

Information Processing Method and Apparatus, Program and
Recording Medium

CROSS-REFERENCE TO RELATED APPLICATIONS

[0001] The present application claims priority from International Application No. PCT/JP01/03417 which was published in Japanese on November 1, 2001, the disclosure of which is hereby incorporated by reference herein.

BACKGROUND OF THE INVENTION

[0002] This invention relates to an information processing method and apparatus, a program, and a recording medium and, more particularly, to an information processing method and apparatus, a program and a recording medium configured for maintaining continuity of moving pictures in a reproducing domain.

[0003] Recently, a variety of types of optical discs have been proposed as a recording medium that can be removed from a recording apparatus. These recordable optical discs have been proposed as a large capacity medium of several GBs and are thought to be promising as a medium for recording AV (audio/video) signals, such as video signals. Among the sources of the digital AV signals (supply sources) to be recorded on this recordable optical disc, there are CS digital satellite broadcasts and BS digital broadcasts. Additionally, ground wave television broadcasts of the digital system have also been proposed for future use.

[0004] The digital video signals, supplied from these sources, are routinely compressed under the MPEG (Motion Picture Experts Group) 2 system. In a recording apparatus, a recording rate proper to the apparatus is set. If digital video signals of the digital broadcast are recorded in the conventional image storage

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mediums for domestic use, digital video signals are first decoded and subsequently bandwidth-limited for recording. In the case of the digital recording system, including, of course, the MPEG1 video, MPEG2 video and DV systems, digital video signals are first decoded and subsequently re-encoded in accordance with an encoding system for the recording rate proper to the apparatus for recording subsequently.

[0005] However, this recording system, in which the supplied bit stream is decoded once and subsequently bandwidth-limited and re-encoded prior to recording, suffers from deteriorated picture quality. If, in recording compressed digital video signals, the transmission rate of input digital signals is less than the recording rate for the recording and/or reproducing apparatus, the method of directly recording the supplied bit stream without decoding or re-encoding suffers from deterioration in the picture quality only to the least extent. However, if the transmission rate of the input digital signals exceeds the recording rate of the recording and/or reproducing apparatus, it is indeed necessary to re-encode the bit stream and to record the re-encoded bit stream so that, after decoding in the recording and/or reproducing apparatus, the transmission rate will be not higher than the upper limit of the disc recording rate.

[0006] If the bit stream is transmitted in a variable rate system in which the bit rate of the input digital signal is increased or decreased with time, the capacity of the recording medium can be exploited less wastefully in the case of a disc recording apparatus adapted for transiently storing data in a buffer and for recording the data in a burst fashion than in the case of a tape recording system having a fixed recording rate

imposed by the fixed rpm of the rotary head.

[0007] Thus, it may be predicted that, in the near future when digital broadcasts are to become the mainstream, an increasing demand will be created for a recording and/or reproducing apparatus in which broadcast signals are recorded as digital signals, without decoding or re-encoding, as in a data streamer, and in which a disc is used as a recording medium.

[0008] In reproducing data recorded on a recording medium in the above-described recording apparatus, there is known so-called skip reproduction in which reproduction is continued up to a preset picture and a picture temporally spaced apart from the preset picture is reproduced next. This skip reproduction has a drawback that temporal continuity is lost in the reproduced pictures.

SUMMARY OF THE INVENTION

[0009] It is therefore an object of the present invention to enable reproduction such as to maintain continuity of moving pictures in the playback domain.

[0010] The present invention provides an information processing apparatus including a generator operable to generate, when continuous reproduction from a first AV stream to a second AV stream is commanded, a third AV stream and address information as information pertinent to the third AV stream, the third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream and being reproduced when reproduction is switched from the first AV stream to the second AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third

AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream, and a recorder operable to record the third AV stream and the address information generated by the generator.

[0011] Preferably, an arrival time stamp of the source packet of the first AV stream is continuous with an arrival time stamp of a source packet located at a leading end of the third AV stream, and an arrival time stamp of the source packet of the second AV stream is continuous with an arrival time stamp of a source packet located at a trailing end of the third AV stream.

[0012] Preferably, a sole discontinuous point exists in the arrival time stamp of the source packet in the third AV stream.

[0013] Preferably, the address is determined so that a data portion of an AV stream previous to a source packet specified by the information on the address of the source packet of the first AV stream is located in a continuous area of not less than a preset size on a recording medium.

[0014] Preferably, the address is determined so that a data portion of an AV stream subsequent to a source packet specified by the information on the address of the source packet of the second AV stream is located in a continuous area of not less than a preset size on a recording medium.

[0015] Preferably, the third AV stream is generated so that the third AV stream will be located in a continuous area of not less than a preset size on the recording medium.

[0016] The present invention also provides an information generating method, including generating, when continuous reproduction from a first AV stream to a second AV stream is

commanded, a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, the third AV stream being reproduced when reproduction is switched from the first AV stream to the second AV stream, and generating address information as information pertinent to the third AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream.

[0017] The present invention also provides a recording medium having recorded thereon a computer-readable program for generating information, in which the program includes generating, when continuous reproduction from a first AV stream to a second AV stream is commanded, a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, the third AV stream being reproduced when reproduction is switched from the first AV stream to the second AV stream, and generating address information as information pertinent to the third AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream.

[0018] The present invention also provides a program for having a computer execute a program for generating information, in which the program includes generating, when continuous

reproduction from a first AV stream to a second AV stream is commanded, a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, the third AV stream being reproduced when reproduction is switched from the first AV stream to the second AV stream, and generating address information as information pertinent to the third AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream.

[0019] The present invention also provides an information processing apparatus, including a reproducing unit operable to reproduce a recording medium having recorded thereon a first AV stream, a second AV stream, a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, and address information as information pertinent to the third AV stream, the third AV stream being reproduced when reproduction is switched from the first AV stream to the second AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream; and a controller operable to control the reproducing unit for switching reproduction from the first AV stream read out from the first AV stream to the third AV stream and from the third AV stream to the

second AV stream, based on the information pertinent to the third AV stream, read out from the second unit.

[0020] The present invention also provides an information processing method, including reproducing a recording medium having recorded thereon a first AV stream, a second AV stream, a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, and address information as information pertinent to the third AV stream, the third AV stream being reproduced when reproduction is switched from the first AV stream to the second AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream; and controlling the reproducing step for switching reproduction from the first AV stream readout-controlled from the first readout controlling step to the third AV stream and from the third AV stream to the second AV stream, based on the information pertinent to the third AV stream, readout-controlled in the second readout controlling step.

[0021] The present invention also provides a recording medium having recorded thereon a computer-readable program for processing information, in which the program includes reproducing a recording medium having recorded thereon a first AV stream, a second AV stream, a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, and address information as information pertinent to the third AV stream, the third AV stream being reproduced when reproduction is

switched from the first AV stream to the second AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream; and controlling the reproducing step for switching reproduction from the first AV stream readout-controlled from the first readout controlling step to the third AV stream and from the third AV stream to the second AV stream, based on the information pertinent to the third AV stream, readout-controlled in the second readout controlling step.

[0022] The present invention also provides a program for having a computer execute a program for processing information, in which the program includes reproducing a recording medium having recorded thereon a first AV stream, a second AV stream, a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, and address information as information pertinent to the third AV stream, the third AV stream being reproduced when reproduction is switched from the first AV stream to the second AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream; and controlling the reproducing step for switching reproduction from the first AV stream readout-controlled from the first readout controlling step to the third

AV stream and from the third AV stream to the second AV stream, based on the information pertinent to the third AV stream, readout-controlled in the second readout controlling step.

[0023] The present invention also provides a recording medium having recorded thereon address information, including, when continuous reproduction from a first AV stream to a second AV stream is commanded, a third AV stream made up of a preset portion of the first AV stream and a preset portion of the second AV stream, the third AV stream being reproduced when reproduction is switched from the first AV stream to the second AV stream; and address information as information pertinent to the third AV stream, the address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream.

[0024] In the information processing method, apparatus, and program according to the present invention, if there is a command to perform reproduction continuously from the first AV stream to the second AV stream, there is generated a third AV stream including a preset portion of the first AV stream and a preset portion of the second AV stream, and the third AV stream is reproduced when reproduction is switched from the first AV stream to the second AV stream, while there is also generated, as information pertinent to the third AV stream, address information including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and information on the

address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream.

[0025] In the information processing method, apparatus, and program according to the present invention, a first AV stream, a second AV stream or a third AV stream is read out from a recording medium; address information, including information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream and information on an address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream, is read out from the recording medium as information pertinent to the third AV stream; and reproduction is switched from the first AV stream to the third AV stream and from the third AV stream to the second AV stream as reproduction proceeds, based on the read-out information pertinent to the third AV stream.

[0026] Other objects, features and advantages of the present invention will become more apparent from reading the embodiments of the present invention as shown in the drawings.

BRIEF DESCRIPTION OF THE DRAWINGS

[0027] Fig. 1 shows a configuration of an embodiment of a recording and/or reproducing apparatus according to the present invention. .

[0028] Fig. 2 illustrates the data format of data recorded on a recording medium by a recording and/or reproducing apparatus 1.

[0029] Fig. 3 illustrates Real PlayList and Virtual PlayList.

[0030] Figs. 4A, 4B and 4C illustrate the creation of the Real PlayList.

- [0031] Figs. 5A, 5B and 5C illustrate deletion of the Real PlayList.
- [0032] Figs. 6A and 6B illustrate assemble editing.
- [0033] Fig. 7 illustrates provision of a sub path in the Virtual PlayList.
- [0034] Fig. 8 illustrates the changing of the playback sequence of the PlayList.
- [0035] Fig. 9 illustrates a mark on the PlayList and a mark on the Clip.
- [0036] Fig. 10 illustrates a menu thumbnail.
- [0037] Fig. 11 illustrates a mark added to the PlayList.
- [0038] Fig. 12 illustrates a mark added to the Clip.
- [0039] Fig. 13 illustrates the relation between the PlayList, Clip and the thumbnail file.
- [0040] Fig. 14 illustrates a directory structure.
- [0041] Fig. 15 illustrates a syntax of infr.dvr.
- [0042] Fig. 16 shows a syntax of DVRVolume.
- [0043] Fig. 17 shows a syntax of ResumeVolume.
- [0044] Fig. 18 shows a syntax of UIAppInfoVolume.
- [0045] Fig. 19 shows a table of character set values.
- [0046] Fig. 20 shows a syntax of TableOfPlayList.
- [0047] Fig. 21 shows another syntax of TableOfPlayList.
- [0048] Fig. 22 shows a syntax of the MakersPrivateData.
- [0049] Fig. 23 shows a syntax of xxxx.rpls and yyyy.vpls.
- [0050] Figs. 24A to 24C illustrate the PlayList.
- [0051] Fig. 25 shows a syntax of PlayList.
- [0052] Fig. 26 shows a table of PlayList_type.
- [0053] Fig. 27 shows a syntax of UIAppInfoPlayList.
- [0054] Figs. 28A to 28C illustrate flags in the

UIAppInfoPlayList syntax shown in Fig. 27.

- [0055] Fig. 29 illustrates a PlayItem.
- [0056] Fig. 30 illustrates a PlayItem.
- [0057] Fig. 31 illustrates a PlayItem.
- [0058] Fig. 32 shows a syntax of the PlayItem.
- [0059] Fig. 33 illustrates IN-time.
- [0060] Fig. 34 illustrates OUT-time.
- [0061] Fig. 35 shows a table of Connection_Condition.
- [0062] Figs. 36A to 36D illustrate Connection_Condition.
- [0063] Fig. 37 illustrates BridgeSequenceInfo.
- [0064] Fig. 38 shows a syntax of BridgeSequenceInfo.
- [0065] Fig. 39 illustrates SubPlayItem.
- [0066] Fig. 40 shows a syntax of SubPlayItem.
- [0067] Fig. 41 shows a table of Mark_type.
- [0068] Fig. 42 shows a syntax of PlaylistMark.
- [0069] Fig. 43 shows a table of Mark_type.
- [0070] Fig. 44 illustrates Mark_time_stamp.
- [0071] Fig. 45 shows a syntax of zzzz.clip.
- [0072] Fig. 46 shows a syntax of ClipInfo.
- [0073] Fig. 47 shows a table of Clip_stream_type.
- [0074] Fig. 48 illustrates offset_SPN.
- [0075] Fig. 49 illustrates offset_SPN.
- [0076] Figs. 50A, 50B illustrate the STC domain.
- [0077] Fig. 51 illustrates STC_Info.
- [0078] Fig. 52 shows a syntax of STC_Info.
- [0079] Fig. 53 illustrates ProgramInfo.
- [0080] Fig. 54 shows a syntax of ProgramInfo.
- [0081] Fig. 55 shows a syntax of VideoCodingInfo.
- [0082] Fig. 56 shows a table of Video_format.

- [0083] Fig. 57 shows a table of frame_rate.
- [0084] Fig. 58 shows a table of display_aspect_ratio.
- [0085] Fig. 59 shows a syntax of AudioCodingInfo.
- [0086] Fig. 60 shows a table of audio_coding.
- [0087] Fig. 61 shows a table of audio_component_type.
- [0088] Fig. 62 shows a table of sampling_frequency.
- [0089] Fig. 63 illustrates CPI.
- [0090] Fig. 64 illustrates CPI.
- [0091] Fig. 65 shows a syntax of CPI.
- [0092] Fig. 66 shows a table of CPI_type.
- [0093] Fig. 67 illustrates video EP_map.
- [0094] Fig. 68 illustrates EP_map.
- [0095] Fig. 69 illustrates EP_map.
- [0096] Fig. 70 shows a syntax of EP_map.
- [0097] Fig. 71 shows a table of EP_typevalues.
- [0098] Fig. 72 shows a syntax of EP_map_for_one_stream_PID.
- [0099] Fig. 73 illustrates TU_map.
- [0100] Fig. 74 shows a syntax of TU_map.
- [0101] Fig. 75 shows a syntax of ClipMark.
- [0102] Fig. 76 shows a table of Mark_type.
- [0103] Fig. 77 shows a table of Mark_type_stamp.
- [0104] Fig. 78 shows a syntax of menu.thmb and mark.thmb.
- [0105] Fig. 79 shows the syntax of thumbnail.
- [0106] Fig. 80 shows a table of thumbnail_picture_format.
- [0107] Figs. 81A and 81B illustrate tn_block.
- [0108] Fig. 82 illustrates a structure of a transport stream of DVR MPEG2.
- [0109] Fig. 83 shows a recorder model of a transport stream of DVR MPEG2.

- [0110] Fig. 84 shows a player model of a transport stream of DVR MPEG2.
- [0111] Fig. 85 shows the syntax of a source packet.
- [0112] Fig. 86 shows the syntax of TP_extra_header.
- [0113] Fig. 87 shows a table of a copy permission indicator.
- [0114] Fig. 88 illustrates seamless connection.
- [0115] Fig. 89 illustrates seamless connection.
- [0116] Fig. 90 illustrates seamless connection.
- [0117] Fig. 91 illustrates seamless connection.
- [0118] Fig. 92 illustrates seamless connection.
- [0119] Fig. 93 illustrates audio overlap.
- [0120] Fig. 94 illustrates seamless connection employing BridgeSequence.
- [0121] Fig. 95 illustrates seamless connection not employing BridgeSequence.
- [0122] Fig. 96 shows a DVR STD model.
- [0123] Fig. 97 is a timing chart for decoding and display.
- [0124] Fig. 98 shows another syntax of BridgeSequenceInfo.
- [0125] Fig. 99 illustrates Bridge_Clip when two PlayItems are seamlessly connected.
- [0126] Fig. 100 shows a syntax of a Clip Information file.
- [0127] Fig. 101 shows a syntax of a ClipInfo of the Clip Information file of Fig. 100.
- [0128] Fig. 102 shows a syntax of a SequenceInfo of the Clip Information file of Fig. 100.
- [0129] Figs. 103A and 103B illustrate database change in case stream data of the Clip AV stream file is partially erased.
- [0130] Fig. 104 is a flowchart for illustrating the preparation of a RealPlayList.

[0131] Fig. 105 is a flowchart for illustrating the formulation of a Virtual PlayList.

[0132] Fig. 106 is a flowchart for illustrating the formulation of a bridge sequence.

[0133] Fig. 107 is a flowchart for illustrating the reproduction of PlayList.

[0134] Fig. 108 illustrates a medium.

DETAILED DESCRIPTION

[0135] Referring to the drawings, an embodiment of the present invention will be explained in detail. Fig. 1 shows a typical inner structure of a recording and/or reproducing apparatus 1 embodying the present invention. First, the structure of a recording unit 2, configured for recording signals input from outside, is explained. The recording and/or reproducing apparatus 1 is configured for being fed with and recording analog or digital data.

[0136] Analog video signals and analog audio signals are fed to terminals 11, 12, respectively. The video signals, input to the terminal 11, are output to an analysis unit 14 and to an AV encoder 15. The audio signals, input to the terminal 12, are output to the analysis unit 14 and to the AV encoder 15. The analysis unit 14 extracts feature points, such as scene changes, from the input video and audio signals.

[0137] The AV encoder 15 encodes input video and audio signals to output the system information (S), such as an encoded video stream (V), an encoded audio stream (A) and AV synchronization, to a multiplexer 16.

[0138] The encoded video stream is a video stream encoded, e.g., with the MPEG (Motion Picture Expert Group) 2 system.

whilst the encoded audio stream is an audio stream encoded in accordance with the MPEG1 system, with the encoded audio stream being, e.g., an audio stream encoded in, e.g., the MPEG1 system or an audio stream encoded in accordance with the Dolby AC3 (trademark) system. The multiplexer 16 multiplexes the input video and audio streams, based on the input system information, to output a multiplexed stream through a switch 17 to a multiplexed stream analysis unit 18 and to a source packetizer 19.

[0139] The multiplexed stream is, e.g., an MPEG-2 transport stream or an MPEG2 program stream. The source packetizer 19 encodes the input multiplexed stream into an AV stream composed of source packets in accordance with an application format of a recording medium 100 on which to record the stream. The AV stream is processed in ECC (error correction and coding) unit 20 and a modulation unit 21 with appendage of ECC codes and with modulation, before being output to a write unit 22, which then writes (records) an AV stream file based on control signals output by the controller 23.

[0140] The transport stream, such as digital television broadcast, input from a digital interface or a digital television tuner, is input to a terminal 13. There are two recording systems for recording the transport stream input to the terminal 13, one being a transparent recording system and the other being a system in which recording is preceded by re-encoding aimed to lower, e.g., the recording bit rate. The recording system command information is input from a terminal 24 as a user interface to the controller 23.

[0141] In the transparent recording of the input transport

stream, a transport stream, input to terminal 13, is output through switch 17 to the multiplexed stream analysis unit 18 and to the source packetizer 19. The ensuing process of recording an AV stream on a recording medium is the same as that of encoding and recording analog input audio and video signals, as described above, and hence is not explained here for simplicity.

[0142] If the input transport stream is re-encoded and subsequently recorded, the transport stream, input to the terminal 13, is fed to a demultiplexer 26, which demultiplexes the input transport stream to extract a video stream (V), an audio stream (A) and the system information (S).

[0143] Of the stream (information), as extracted by the demultiplexer 26, the video stream is output to an audio decoder 27, whilst the audio stream and the system information are output to the multiplexer 16. The audio decoder 27 decodes the input transport stream to output the encoded video stream (V) to the multiplexer 16.

[0144] The audio stream and the system information, output from the demultiplexer 26 and input to the multiplexer 16, and the video stream, output by the AV encoder 15, are multiplexed, based on the input system information, and output to the multiplexed stream analysis unit 18 and to the source packetizer 19 through switch 17, as a multiplexed stream. The ensuing process of recording an AV stream on a recording medium is the same as that of encoding and recording analog input audio and video signals, as described above, and hence is not explained here for simplicity.

[0145] The recording and/or reproducing apparatus 1 of the present embodiment records a file of the AV stream on the

recording medium 100, while also recording the application database information which accounts for the file. The input information to the controller 23 is the feature information for the moving picture from the analysis unit 14, the feature information of the AV stream from the multiplexed stream analysis unit 18 and the user command information input at terminal 24.

[0146] The feature information of the moving picture, supplied from the analysis unit 14, is generated by the analysis unit 14 when the AV encoder 15 encodes video signals. The analysis unit 14 analyzes the contents of the input video and audio signals to generate the information pertinent to the pictures characteristic of the input moving picture signals (clip mark). This information is the information indicating a picture of characteristic clip mark points, such as program start points, scene change points, CM commercial start and end points, title or telop in input video signals, and also includes a thumbnail of the picture and the information pertinent to stereo/monaural switching points and muted portions of audio signals.

[0147] The above picture indicating information is fed through controller 23 to the multiplexer 16. When multiplexing an encoded picture specified as clip mark by the controller 23, the multiplexer 16 returns the information for specifying the encoded picture on the AV stream to the controller 23. Specifically, this information is the PTS (presentation time stamp) of a picture or the address information on the AV stream of an encoded version of the picture. The controller 23 stores the sort of feature pictures and the information for specifying the encoded picture on the AV stream in association with each other.

[0148] The feature information of the AV stream from the

multiplexed stream analysis unit 18 is the information pertinent to the encoding information of the AV stream to be recorded, and is recorded by the analysis unit 18. For example, the feature information includes the time stamp and address information of the I-picture in the AV stream, discontinuous point information of system time clocks, encoding parameters of the AV stream and change point information of the encoding parameters in the AV stream. When transparently recording the transport stream input from the terminal 13, the multiplexed stream analysis unit 18 detects the picture of the aforementioned clip mark from the input transport stream, and generates the information for specifying a picture designated by the clip mark and its type.

[0149] The user designation information from the terminal 24 is the information specifying the playback domain, designated by the user, character letters for explaining the contents of the playback domain, or the information such as bookmarks or resuming points set by the user for his or her favorite scene.

[0150] Based on the aforementioned input information, the controller 23 creates a database of the AV stream (Clip), a database of a group (PlayList) of playback domains (PlayItem) of the AV stream, management information of the recorded contents of the recording medium 100 (info.dvr) and the information on thumbnail pictures. Similarly to the AV stream, the application database information, constructed from the above information, is processed in the ECC unit 20 and the modulation unit 21 and input to the write unit 22, which then records a database file on the recording medium 100.

[0151] The above-described application database information will be explained subsequently in detail.

[0152] When the AV stream file recorded on the recording medium 100 (files of picture data and speech data) and the application database information, thus recorded on the recording medium 100, are reproduced by a reproducing unit 3, the controller 23 first commands a readout unit 28 to read out the application database information from the recording medium 100. The readout unit 28 reads out the application database information from the recording medium 100, and sends the application database information through demodulation and error correction processing by a demodulating unit 29 and an ECC decoder 30 to the controller 23.

[0153] Based on the application database information, the controller 23 outputs a list of PlayList recorded on the recording medium 100 to a user interface of the terminal 24. The user selects the PlayList desired to be reproduced from the list of PlayLists. The information pertinent to PlayList, specified to be reproduced, is input to the controller 23. The controller 23 commands the readout unit 28 to read out the AV stream file necessary in reproducing the PlayList. In accordance with the command, the readout unit 28 reads out the corresponding AV stream from the recording medium 100 to output the read-out AV stream to the demodulating unit 29. The AV stream, thus input to the demodulating unit 29, is demodulated by preset processing and output through the processing by the ECC decoder 30 to a source depacketizer 31.

[0154] The source depacketizer 31 converts the AV stream of the application format, read out from the recording medium 100 and processed in a preset fashion, into a stream processable by the demultiplexer 26. The demultiplexer 26 outputs the system

information (S), such as the video stream (V), audio stream (A) or the AV synchronization, forming the playback domain (PlayItem) of the AV stream specified by the controller 23, to the audio decoder 27, which AV decoder 27 decodes the video stream and the audio stream to output the playback video signal and the playback audio signal to associated terminals 32, 33, respectively.

[0155] If fed from the terminal 24, as a user interface, with information instructing random access playback or special playback, the controller 23 determines the readout position of the AV stream from the recording medium 100, based on the contents of the database (Clip) of the AV stream, to command the readout unit 28 to read out the AV stream. If the PlayList as selected by the user is to be reproduced as from a preset time point, the controller 23 commands the readout unit 28 to read out data from an I-picture having a time stamp closest to the specified time point.

[0156] When the user has selected a certain clip mark from indexing points or scene change points for the program stored in the ClipMark in the Clip Information, as when the user selects a certain picture from a list of thumbnail pictures, as demonstrated on a user interface, of the indexing points or scene change points stored in the ClipMark, the controller 23 determines the AV stream readout position from the recording medium 100 to command the readout unit 28 to read out the AV stream. That is, the controller 23 commands the readout unit 28 to read out data from an I-picture having an address closest to the address on the AV stream which has stored the picture selected by the user. The readout unit 28 reads out data from the specified address. The read-out data is processed by the

demodulating unit 29, ECC decoder 30 and by the source packetizer 19 so as to be supplied to the demultiplexer 26 and decoded by the audio decoder 27 to reproduce AV data indicated by an address of the mark point picture.

[0157] If the user has commanded fast forward playback, the controller 23 commands the readout unit 28 to sequentially read out I-picture data in the AV stream in succession based on the database (Clip) of the AV stream.

[0158] The readout unit 28 reads out data of the AV stream from a specified random access point. The so read-out data is reproduced through processing by various components on the downstream side.

[0159] The case in which the user edits the AV stream recorded on the recording medium 100 is now explained. If desired to specify a playback domain for the AV stream recorded on the recording medium 100, for example, if desired to create a playback route of reproducing a portion sung by a singer A from a song program A, and subsequently reproducing a portion sung by the same singer A from another song program B, the information pertinent to a beginning point (IN-point) and an end point (OUT-point) of the playback domain is input to the controller 23 from the terminal as a user interface. The controller 23 creates a database of the group (PlayList) of playback domains (PlayItem) of the AV streams.

[0160] When the user desires to erase a portion of the AV stream recorded on the recording medium 100, the information pertinent to the IN-point and the OUT-point of the erasure domain is input to the controller 23, which then modifies the database of the PlayList so as to refer to only the needed AV streams.

The controller 23 also commands the write unit 22 to erase an unneeded stream portion of the AV stream.

[0161] The case in which the user desires to specify playback domains of an AV stream recorded on the recording medium to create a new playback route, and to interconnect the respective playback domains in a seamless fashion, is now explained. In such case, the controller 23 creates a database of a group (PlayList) of the playback domains (PlayItem) of the AV stream and undertakes to partially re-encode and re-multiplex the video stream in the vicinity of junction points of the playback domains.

[0162] The picture information at the IN-point and that at the OUT-point of a playback domain are input from a terminal 24 to a controller 23. The controller 23 commands the readout unit 28 to read out data needed to reproduce the pictures at the IN-point and at the OUT-point. The readout unit 28 reads out data from the recording medium 100. The data so read out is output through the demodulating unit 29, ECC decoder 30 and the source packetizer 19 to the demultiplexer 26.

[0163] The controller 23 analyzes data input to the demultiplexer 26 to determine the re-encoding method for the video stream (change of picture_coding_type and assignment of the quantity of encoding bits for re-encoding) and the re-multiplexing system to send the system to the AV encoder 15 and to the multiplexer 16.

[0164] The demultiplexer 26 then separates the input stream into the video stream (V), audio stream (A) and the system information (S). The video stream may be classed into data input to the audio decoder 27 and data input to the multiplexer 16.

The former is data needed for re-encoding, and is decoded by the audio decoder 27, with the decoded picture being then re-encoded by the AV encoder 15 and thereby caused to become a video stream. The latter data is data copied from an original stream without re-encoding. The audio stream and the system information are directly input to the multiplexer 16.

[0165] The multiplexer 16 multiplexes an input stream, based on the information input from the controller 23, to output a multiplexed stream, which is processed by the ECC unit 20 and the modulation unit 21 so as to be sent to the write unit 22. The write unit 22 records an AV stream on the recording medium 100 based on the control signals supplied from the controller 23.

[0166] The application database information and the operation based on this information, such as playback and editing, are hereinafter explained. Fig. 2 shows the structure of an application format having two layers, that is PlayList and Clip, for AV stream management. The Volume Information manages all Clips and PlayLists in the disc. Here, one AV stream and the ancillary information thereof, paired together, is deemed to be an object, and is termed Clip. The AV stream file is termed a Clip AV stream file, with the ancillary information being termed the Clip Information file.

[0167] One Clip AV stream file stores data corresponding to an MPEG-2 transport stream arranged in a structure prescribed by the application format. By and large, a file is treated as a byte string. The contents of the Clip AV stream file are expanded on the time axis, with entry points in the Clip (I-picture) being mainly specified on the time basis. When a time stamp of an access point to a preset Clip is given, the Clip Information file

is useful in finding the address information at which to start data readout in the Clip AV stream file.

[0168] Referring to Fig. 3, PlayList is now explained, which is provided for a user to select a playback domain desired to be viewed from the Clip and to edit the playback domain readily. One PlayList is a set of playback domains in the Clip. One playback domain in a preset Clip is termed PlayItem and is represented by a pair of an IN-point and an OUT-point on the time axis. So, the PlayList is formed by a set of plural PlayItems.

[0169] The PlayList is classified into two types, one of which is Real PlayList and the other of which is Virtual PlayList. The Real PlayList co-owns stream portions of the Clip it is referencing. That is, the Real PlayList takes up in the disc the data capacity corresponding to a stream portion of the Clip it is referencing and, when Real PlayList is erased, the data of the stream portion of the Clip it is referencing is also erased.

[0170] The Virtual PlayList is not co-owning Clip data. Therefore, if the Virtual PlayList is changed or erased, the contents of the Clip are in no way changed.

[0171] The editing of the Real Playlist is explained. Fig. 4A shows creation of Real PlayList and, if the AV stream is recorded as a new Clip, the Real PlayList which references the entire Clip is a newly created operation.

[0172] Fig. 4B shows the division of the Real PlayList, that is the operation of dividing the Real PlayList at a desired point to split the Real PlayList into two Real PlayLists. This division operation is performed when two programs are managed in one clip managed by a sole PlayList and when the user intends to re-register or re-record the programs as separate individual

programs. This operation does not lead to alteration of the Clip contents, that is, to division of the Clip itself.

[0173] Fig. 4C shows the operation of combining the Real PlayList which is the operation of combining two Real PlayLists into one new Real PlayList. This combining operation is performed such as when the user desires to re-register two programs as a sole program. This operation does not lead to alteration of the Clip contents, that is, to combining the Clip itself into one.

[0174] Fig. 5A shows deletion of the entire Real PlayList. In the operation of erasing the entire preset Real PlayList, the associated stream portion of the Clip referenced by the deleted Real PlayList is also deleted.

[0175] Fig. 5B shows partial deletion of the Real PlayList. If a desired portion of the Real PlayList is deleted, the associated PlayItem is altered to reference only the needed Clip stream portion. The corresponding stream portion of the Clip is deleted.

[0176] Fig. 5C shows the minimizing of the Real Playlist. This is an operation of causing the PlayItem associated with the Real Playlist to reference only the stream portion of the Clip needed for Virtual Playlist. The corresponding stream portion of the Clip not needed for the Virtual Playlist is deleted.

[0177] If the Real PlayList is changed by the above-described operation such that the stream portion of the Clip referenced by the Real PlayList is deleted, there is a possibility that the Virtual PlayList employing the deleted Clip is present such that problems may be produced in the Virtual PlayList due to the deleted Clip.

[0178] In order to prevent this from occurring, such a message which runs: "If there exists the Virtual PlayList referencing the stream portion of the Clip the Real PlayList is referencing, and the Real PlayList is deleted, the Virtual PlayList itself is deleted - is it all right?" is displayed for the user in response to the user's operation of deletion by way of confirmation or alarming, after which the processing for deletion is executed or cancelled subject to the user's commands. Alternatively, the minimizing operation for the Real PlayList is performed in place of deleting the Virtual PlayList.

[0179] The operation for the Virtual PlayList is now explained. If an operation is performed for the Virtual PlayList, the contents of the Clip are not changed. Figs. 6A and 6B show the assembling and editing (IN-OUT editing). This is an operation of creating PlayItem of the playback domain the user has desired to view to create Virtual PlayList. The seamless connection between PlayItems is supported by the application format, as later explained.

[0180] If there exist two Real PlayLists 1, 2 and clips 1, 2 associated with the respective Real PlayLists, the user specifies a preset domain in the Real PlayList 1 (domain from IN1 to OUT1: PlayItem 1) as the playback domain, and also specifies, as the domain to be reproduced next, a preset domain in the Real PlayList 2 (domain from IN2 to OUT2: PlayItem 2) as the playback domain, as shown in Fig. 6A, and a sole Virtual PlayList made up of PlayItem 1 and the PlayItem2 is prepared, as shown in Fig. 6B.

[0181] The re-editing of the Virtual PlayList is now explained. The re-editing may be enumerated by alteration of IN or OUT points in the Virtual PlayList, insertion or appendage of

new PlayItems to the Virtual PlayList and deletion of PlayItems in the Virtual PlayList. The Virtual PlayList itself may also be deleted.

[0182] Fig. 7 shows the audio dubbing (post recording) to the Virtual PlayList. This is an operation of registering the audio post recording to the Virtual PlayList as a sub path. This audio post recording is supported by the application software. An additional audio stream is added as a sub path to the AV stream of the main path of the Virtual PlayList.

[0183] Common to the Real PlayList and the Virtual PlayList is an operation of changing (moving) the playback sequence of the PlayList shown in Fig. 8. This operation is an alteration of the playback sequence of the PlayList in the disc (volume) and is supported by TableOfPlayList as defined in the application format, as will be explained subsequently with reference to, e.g., Fig. 20. This operation does not lead to alteration of the Clip contents.

[0184] The mark (Mark) is now explained. The mark is provided for specifying a highlight or characteristic time in the Clip and in the PlayList, as shown in Fig. 9. The mark added to the Clip is termed the ClipMark. The ClipMark is, e.g., a program indexing point or a scene change point for specifying a characteristic scene ascribable to contents in the AV stream. The ClipMark is generated by, e.g., the analysis unit 14 of Fig. 1. When the PlayList is reproduced, the mark of the Clip referenced by the PlayList may be referenced and used.

[0185] The mark appended to the PlayList is termed the PlayListMark (play list mark). The PlayListMark is, e.g., a bookmark point or a resuming point as set by the user. The mark

is set to the Clip and to the PlayList by adding a time stamp indicating the mark time point to the mark list. On the other hand, mark deletion is removing the time stamp of the mark from the mark list. Consequently, the AV stream is in no way changed by mark setting or by mark deletion.

[0186] As another format of the ClipMark, a picture referenced by the ClipMark may be specified on the address basis in the AV stream. Mark setting on the Clip is by adding the address basis information indicating the picture of the mark point to the mark list. On the other hand, mark deletion is removing the address basis information indicating the mark point picture from the mark list. Consequently, the AV stream is in no way changed by mark setting or by mark deletion.

[0187] A thumbnail is now explained. The thumbnail is a still picture added to the Volume, PlayList and Clip. There are two sorts of the thumbnail, one of them being a thumbnail as a representative picture indicating the contents. This is mainly used in a main picture in order for the user to select what he or she desired to view on acting on a cursor, not shown. Another sort of the thumbnail is a picture indicating a scene pointed to by the mark.

[0188] The Volume and the respective PlayLists need to own representative pictures. The representative pictures of the Volume are presupposed to be used for initially demonstrating a still picture representing the disc contents when the disc is set in position in the recording and/or reproducing apparatus 1. It is noted that the disc means the recording medium 100 which is presupposed to be of a disc shape. The representative picture of the PlayList is presupposed to be used as a still picture for

representing PlayList contents.

[0189] As the representative picture of the PlayList, it may be contemplated to use the initial picture of the PlayList as the thumbnail (representative picture). However, the leading picture at the playback time of 0 is not necessarily an optimum picture representing the contents. So, the user is allowed to set an optional picture as a thumbnail of the PlayList. Two sorts of the thumbnails, that is, the thumbnail as a representative picture indicating the Volume and the thumbnail as a representative picture indicating PlayList, are termed menu thumbnails. Since the menu thumbnails are demonstrated frequently, these thumbnails need to be read out at an elevated speed from the disc. Thus, it is efficient to store the totality of the menu thumbnails in a sole file. It is unnecessary for the menu thumbnails to be pictures extracted from the moving pictures in the Volume, but may be a picture captured from a personal computer or a digital still camera, as shown in Fig. 10.

[0190] On the other hand, the Clip and the PlayList need to be marked with plural marks, whilst the pictures of the mark points need to be readily viewed in order to grasp the contents of the mark positions. The picture indicating such mark point is termed a mark thumbnail. Therefore, the picture which is the original of the mark thumbnail is mainly an extracted mark point picture rather than a picture captured from outside.

[0191] Fig. 11 shows the relation between the mark affixed to the PlayList and the mark thumbnail, whilst Fig. 12 shows the relation between the mark affixed to the Clip and the mark thumbnail. In distinction from the menu thumbnail, the mark thumbnail is used in, e.g., a sub-menu for representing details

of the PlayList, while it is not requested to be read out in a short access time. So, whenever a thumbnail is required, the recording and/or reproducing apparatus 1 opens a file and reads out a portion of the file, while there is no problem presented even if file opening and reading out a portion of the file by the recording and/or reproducing apparatus 1 takes some time.

[0192] For decreasing the number of files present in a volume, it is preferred for the totality of the mark thumbnails to be stored in one file. While the PlayList may own one menu thumbnail and plural mark thumbnails, the user is not required to select the Clip directly (usually, the Clip is selected through PlayList), and hence there is no necessity of providing menu thumbnails.

[0193] Fig. 13 shows the relation between the menu thumbnails, mark thumbnails, PlayList and Clips. In the menu thumbnail file are filed menu thumbnails provided from one PlayList to another. In the menu thumbnail file is contained a volume thumbnail representing the contents of data recorded on the disc. In the menu thumbnail file are filed thumbnails created from one PlayList to another and from one Clip to another.

[0194] The CPI (Characteristic Point Information) is hereinafter explained. The CPI is data contained in the Clip information file and is used mainly for finding a data address in the Clip AV stream file at which to start the data readout when a time stamp of the access point to the Clip is afforded. In the present embodiment, two sorts of the CPI are used, one of them being EP_map and the other being TU_map.

[0195] The EP_map is a list of entry point (EP) data extracted

from the elementary stream and the transport stream. This has the address information used to find the site of entry points in the AV stream at which to start the decoding. One EP data is made up of a presentation time stamp (PTS) and a data address in the AV stream of the accessing unit associated with the PTS, with the data address being paired to the PTS.

[0196] The EP_map is used mainly for two purposes. First, it is used for finding a data address in the AV stream in the accessing unit referenced by the PTS in the PlayList. Second, the EP_map is used for fast forward playback or fast reverse playback. If, in recording the input AV stream by the recording and/or reproducing apparatus 1, the syntax of the stream can be analyzed, the EP_map is created and recorded on the disc.

[0197] The TU_map has a list of time unit (TU) data which is derived from the arrival time point of the transport packet input through a digital interface. This affords the relation between the arrival-time-based time and the data address in the AV stream. When the recording and/or reproducing apparatus 1 records an input AV stream, and the syntax of the stream cannot be analyzed, a TU_map is created and recorded on the disc.

[0198] The STCInfo stores the discontinuous point information in the AV stream file which stores the MPEG-2 transport stream.

[0199] When the AV stream has discontinuous points of STC, the same PTS values may appear in the AV stream file. Thus, if a time point in the AV stream is specified on the PTS basis, the PTS of the access point is insufficient to specify the point. Moreover, an index of the continuous STC domain containing the PTS is required. In this format, the continuous STC domain and its index are termed an STC-sequence and STC-sequence-id,

respectively. The STC-sequence information is defined by the STCInfo of the Clip Information file.

[0200] The STC-sequence-id is used in an AV stream file and is optional in the AV stream file having the TU_map.

[0201] The programs are each a collection of elementary streams and co-own a sole system time base for synchronized reproduction of these streams.

[0202] It is useful for a reproducing apparatus (recording and/or reproducing apparatus 1 of Fig. 1) to know the contents of an AV stream prior to its decoding. These contents include, e.g., values of the PID of a transport packet transmitting an audio or video elementary stream or the type of the video or audio components, such as HDTV video or MPEG-2 AAC audio stream. This information is useful for creating a menu screen for illustrating to the user the contents of the PlayList referencing the AV stream. It is similarly useful for setting the initial state of the AV decoder and the demultiplexer of the respective apparatus. For this reason, the Clip Information file owns ProgramInfo for illustrating the program contents.

[0203] It may be an occurrence that program contents are changed in the AV stream file in which the MPEG-2 transport stream is stored. For example, the PID of the transport packet transmitting the video elementary stream may be changed, or the component type of the video stream may be changed from SDTV to HDTV.

[0204] The ProgramInfo stores the information on change points of program contents in the AV stream file. The domain of the AV stream file in which the program contents remain constant is termed program-sequence. This program-sequence is used in an AV

stream file having EP_map and is optional in an AV stream file having TU_map.

[0205] The present embodiment defines the self-encoding stream format (SESF). The SESF is used for encoding analog input signals and for decoding digital input signals for subsequently encoding the decoded signal into an MPEG-2 transport stream.

[0206] The SESF defines an elementary stream pertinent to the MPEG-2 transport stream and the AV stream. When the recording and/or reproducing apparatus 1 encodes and records input signals in the SESF, an EP_map is created and recorded on the disc.

[0207] A digital broadcast stream uses one of the following systems for recording on the recording medium 100: first, the digital broadcast stream is transcoded into an SESF stream. In this case, the recorded stream must conform to SESF and EP_map must be prepared and recorded on the disc.

[0208] Alternatively, an elementary stream forming a digital broadcast stream is transcoded to a new elementary stream and remultiplexed to a new transport stream conforming to the stream format prescribed by the organization for standardizing the digital broadcast stream. In this case, an EP_map must be created and recorded on the disc.

[0209] For example, it is assumed that the input stream is an MPEG-2 transport stream conforming to the ISDB (standard appellation of digital BS of Japan), with the transport stream containing the HDTV video stream and the MPEG AAC audio stream. The HDTV video stream is transcoded to an SDTV video stream, which SDTV video stream and the original AAC audio stream are remultiplexed to TS. The SDTV stream and the transport stream both need to conform to the ISDB format.

[0210] Another system of recording the digital broadcast stream on the recording medium 100 is to make transparent recording of the input transport stream, that is to record the input transport stream unchanged, in which case the EP_map is formulated and recorded on the disc.

[0211] Alternatively, the input transport stream is recorded transparently, that is, an input transport stream is recorded unchanged, in which case TU_map is created and recorded on the disc.

[0212] The directory and the file are hereinafter explained. The recording and/or reproducing apparatus 1 is hereinafter described as DVR (digital video recording). Fig. 14 shows a typical directory structure on the disc. The directories of the disc of the DVR may be enumerated by a root directory including "DVR" directory, and the "DVR" directory, including "PLAYLIST" directory, "CLIPINF" directory, "M2TS" directory and "DATA" directory, as shown in Fig. 14. Although directories other than these may be created below the root directory, these are discounted in the application format of the present embodiment.

[0213] Below the "DATA" directory, there are stored all files and directories prescribed by the DVR application format. The "DVR" directory includes four directories. Below the "PLAYLIST" directory are placed Real PlayList and Virtual PlayList database files. The latter directory may exist in a state devoid of PlayList.

[0214] Below "CLIPINF" is placed a Clip database. This directory, too, may exist in a state devoid of AV stream files. In the "DATA" directory, there are stored files of data broadcasts, such as digital TV broadcasts.

[0215] The "DVR" directory stores the following files: an "info.dvr" is created below the DVR directory to store the comprehensive information of an application layer. Below the DVR directory, there must be a sole info.dvr. The filename is assumed to be fixed to info.dvr. The "menu.thmb" stores the information pertinent to the menu thumbnails. Below the DVR directory, there must be 0 or 1 menu thumbnail. The filename is assumed to be fixed to "menu.thmb". If there is no menu thumbnail, this file may not exist.

[0216] The "mark.thmb" file stores the information pertinent to the mark thumbnail picture. Below the DVR directory, there must be 0 or 1 mark thumbnail. The filename is assumed to be fixed to "mark.thmb". If there is no mark thumbnail, this file may not exist.

[0217] The "PLAYLIST" directory stores two sorts of the PlayList files which are Real PlayList and Virtual PlayList. An "xxxxxx.rpls" file stores the information pertinent to one Real PlayList. One file is created for each Real PlayList. The filename is "xxxxxx.rpls", where "xxxxxx" denotes five numerical figures from 0 to 9. A file extender must be "rpls".

[0218] The "yyyyy.vpls" stores the information pertinent to one Virtual PlayList. One file with a filename "yyyyy.vpls" is created from one Virtual PlayList to another, where "yyyyy" denotes five numerical figures from 0 to 9. A file extender must be "vpls".

[0219] The "CLIPINF" directory stores one file in association with each AV stream file. The "zzzzz.clpi" is a Clip Information file corresponding to one AV stream file (Clip AV stream file or Bridge-Clip stream file). The filename is "zzzzz.clpi", where

"zzzzz" denotes five numerical figures from 0 to 9. A file extender must be "clpi".

[0220] The "M2TS" directory stores an AV stream file. The "zzzzz.m2ts" file is an AV stream file handled by the DVR system. This is a Clip AV stream file or a Bridge-Clip AV stream file. The filename is "zzzzz.m2ts", where "zzzzz" denotes five numerical figures from 0 to 9. A file extender must be "m2ts".

[0221] The "DATA" directory stores data transmitted from data broadcasting. This data may, for example, be XML or MPEG files.

[0222] The syntax and the semantics of each directory (file) are now explained. Fig. 15 shows the syntax of the "info.dvr" file. The "info.dvr" file is made up of three objects, that is DVRVolume(), TableOfPlayLists() and MakersPrivateData().

[0223] The syntax of info.dvr shown in Fig. 15 is explained. The TableOfPlayLists_Start_address indicates the leading address of the TableOfPlayLists() in terms of the relative number of bytes from the leading byte of the "info.dvr" file. The relative number of bytes is counted beginning from 0.

[0224] The MakersPrivateData_Start_address indicates the leading address of the MakersPrivateData(), in terms of the relative number of bytes as from the leading byte of the "info.dvr" file. The relative number of bytes is counted from 0. The padding_word is inserted in association with the syntax of "info.dvr". N1 and N2 are optional positive integers. Each padding word may assume an optional value.

[0225] The DVRVolume() stores the information stating the contents of the volume (disc). The syntax of the DVRVolume(), shown in Fig. 16, is now explained. The version_number indicates four character letters indicating the version numbers of the

DVRVolume(). The version_number is encoded to "0045" in association with ISO646.

[0226] Length is denoted by 32-bit unsigned integers indicating the number of bytes from directly after the length field to the trailing end of DVRVolume().

[0227] The ResumeVolume() memorizes the filename of the Real PlayList or the Virtual PlayList reproduced last in the Volume. However, the playback position when the user has interrupted playback of the Real PlayList or the Virtual PlayList is stored in the resume-mark defined in the PlayListMark() (see Figs. 42 and 43).

[0228] The syntax of the ResumeVolume(), shown in Fig. 17, is explained. The valid_flag indicates that the resume_PlayList_name field is valid or invalid when this 1-bit flag is set to 1 or 0, respectively. The 10-byte field of resume_PlayList_name indicates the filename of the Real PlayList or the Virtual PlayList to be resumed.

[0229] The UIAppInfoVolume() in the syntax of the DVRVolume(), shown in Fig. 16, stores parameters of the user interface application concerning the Volume. Fig. 18 shows the syntax of the UIAppInfoVolume(), the semantics of which are now explained. The 8-bit field of character_set indicates the encoding method for character letters encoded in the Volume_name field. The encoding method corresponds to the values shown in Fig. 19.

[0230] The 8-bit field of the name_length indicates the byte length of the Volume name indicated in the Volume_name field. The Volume_name field indicates the appellation of the Volume. The number of bytes of the number of the name_length counted from left of the field is the number of valid characters and indicates

the volume appellation. The values next following these valid character letters may be any values.

[0231] The Volume_protect_flag is a flag indicating whether or not the contents in the Volume can be shown to the user without limitations. If this flag is set to 1, the contents of the Volume are allowed to be presented (reproduced) to the user only in case the user has succeeded in correctly inputting the PIN number (password). If this flag is set to 0, the contents of the Volume are allowed to be presented to the user even in case the PIN number is not input by the user.

[0232] If, when the user has inserted a disc into a player, this flag has been set to 0, or the flag is set to 1 but the user has succeeded in correctly inputting the PIN number, the recording and/or reproducing apparatus 1 demonstrates a list of the PlayList in the disc. The limitations on reproduction of the respective PlayLists are irrelevant to the Volume_protect_flag and are indicated by playback_control_flag defined in the UIAppInfoVolume().

[0233] The PIN is made up of four numerical figures of from 0 to 9, each of which is coded in accordance with ISO/IEC 646. The ref_thumbnail_index field indicates the information of a thumbnail picture added to the Volume. If the ref_thumbnail_index field is of a value other than 0xFFFF, a thumbnail picture is added to the Volume. The thumbnail picture is stored in a menu.thmb file. The picture is referenced using the value of the ref_thumbnail_index in the menu.thmb file. If the ref_thumbnail_index field is 0xFFFF, it indicates that a thumbnail picture has been added to the Volume.

[0234] The TableOfPlayList() in the info.dvr syntax shown in

Fig. 15 is explained. The TableOfPlayList() stores the filename of the PlayList (Real PlayList and Virtual PlayList). All PlayList files recorded in the Volume are contained in the TableOfPlayList(), which TableOfPlayList() indicates the playback sequence of the default of the PlayList in the Volume.

[0235] Fig. 20 shows the syntax of the TableOfPlayList(), which is now explained. The version_number of the TableOfPlayList() indicates four character letters indicating the version numbers of the TableOfPlayLists. The version_number must be encoded to "0045" in accordance with ISO 646.

[0236] Length is an unsigned 32-bit integer indicating the number of bytes of the TableOfPlayList() from directly after the length field to the trailing end of the TableOfPlayList(). The 16-bit field of the number_of_PlayLists indicates the number of loops of the for-loop inclusive of the PlayList_file_name. This numerical figure must be equal to the number of PlayLists recorded in the Volume. The 10-byte numerical figure of the PlayList_file_name indicates the filename of the PlayLists.

[0237] Fig. 21 shows another configuration of the syntax of the TableOfPlayList(). The syntax shown in Fig. 21 is comprised of the syntax shown in Fig. 20 in which is contained the UIAppInfoPlayList. By such structure including the UIAppInfoPlayList, it becomes possible to create a menu picture simply on reading out the TableOfPlayLists. The following explanation is premised on the use of the syntax shown in Fig. 20.

[0238] The MakersPrivateData in the info.dvr shown in Fig. 15 is explained. The MakersPrivateData is provided to permit the maker of the recording and/or reproducing apparatus 1 to insert

private data of the maker in the MakersPrivateData() for special applications of different companies. The private data of each maker has standardized maker_ID for identifying the maker who has defined it. The MakersPrivateData() may contain one or more maker_ID.

[0239] If a preset maker intends to insert private data, and the private data of a different maker is already contained in the MakersPrivateData(), the new private data is added to the MakersPrivateData() without erasing the pre-existing old private data. Thus, in the present embodiment, private data of plural makers can be contained in one MakersPrivateData().

[0240] Fig. 22 shows the syntax of the MakersPrivateData, which is now explained. The version_number of the TableOfPlayList() indicates four character letters indicating the version numbers of the TableOfPlayLists. The version_number must be encoded to "0045" in accordance with ISO 646. Length is an unsigned 32-bit integer indicating the number of bytes of the TableOfPlayList() from directly after the length field to the trailing end of the MakersPrivateData().

[0241] The mpd_blocks_start_address indicates the leading end address of the first mpd_block() in terms of the relative number of bytes from the leading byte of the MakersPrivateData(). The number_of_maker_entries is the 16-bit codeless integer affording the number of entries of the maker private data included in the MakersPrivateData(). There must not be present two or more maker private data having the same maker_ID values in the MakersPrivateData().

[0242] The mpd_blocks_size is a 16-bit unsigned integer affording one mpd_block size in terms of 1024 bytes as a unit.

For example, if `mpd_block_size` = 1, it indicates that the size of one `mpd_block` is 1024 bytes. The `number_of_mpd_blocks` is a 16-bit unsigned integer affording the number of `mpd_blocks` contained in the `MakersPrivateData()`. The `maker_ID` is the 16-bit unsigned integer indicating the model number code of the DVR system which has created the maker private data. The value encoded to the `maker_ID` is specified by the licensor.

[0243] The maker_model_code is a 16-bit unsigned integer indicating the model number code of the DVR system which has created the maker private data. The value encoded to the maker_model_code is set by the maker who has received the license of the format. The start_mpd_block_number is a 16-bit unsigned integer indicating the number of the mpd_block number at which begins the maker private data. The leading end of the maker private data must be aligned with the leading end of the mpd_block. The start_mpd_block_number corresponds to a variable j in the for-loop of the mpd block.

[0244] The `mpd_length` is a 32-bit unsigned integer indicating the size of the maker private data. The `mpd_block` is an area in which is stored maker private data. All of the `mpd_blocks` in the `MakersPrivateData()` must be of the same size.

[0245] The real PlayList file and the Virtual PlayList file, in other words, `xxxxxx.rpls` and `yyyyyy.vpls`, are explained. Fig. 23 shows the syntax of `xxxxxx.rpls` (Real PlayList) and `yyyyyy.vpls` (Virtual PlayList), which are of the same syntax structure. Each of the `xxxxxx.rpls` and `yyyyyy.vpls` is made up of three objects, that is `PlayList()`, `PlayListMark()` and `MakersPrivateData()`.

[0246] The PlayListMark_Start_address indicates the leading address of the PlayListMark(), in terms of the relative number of

bytes from the leading end of the PlayList file as a unit. The relative number of bytes is counted from zero.

[0247] The MakersPrivateData_Start_address indicates the leading address of the MakersPrivateData(), in terms of the relative number of bytes from the leading end of the PlayList file as a unit. The relative number of bytes is counted from zero.

[0248] The padding_word (padding word) is inserted in accordance with the syntax of the PlayList file, with N1 and N2 being optional positive integers. Each padding word may assume an optional value.

[0249] PlayList will be further explained in the following, although it has been explained briefly. A playback domain in all Clips except Bridge-Clip must be referred to by all PlayLists in the disc. Also, two or more Real PlayLists must not overlap the playback domains shown by their PlayItems in the same Clip.

[0250] Reference is made to Figs. 24A, 24B and 24C. For all Clips, there exist corresponding Real PlayLists, as shown in Fig. 24A. This rule is observed even after the editing operation has come to a close, as shown in Fig. 24B. Therefore, all Clips must be viewed by referencing one of Real PlayLists.

[0251] Referring to Fig. 24C, the playback domain of the Virtual PlayList must be contained in the playback domain and in the Bridge-Clip playback domain. There must not be present in the disc Bridge-Clip not referenced by any Virtual PlayList.

[0252] The Real PlayList, containing the list of the PlayItem, must not contain SubPlayItem. The Virtual PlayList contains the PlayItem list and, if the CPI_type contained in the PlayList() is the EP_map type and the PlayList_type is 0 (PlayList containing

video and audio), the Virtual PlayList may contain one SubPlayItem. In the PlayList() in the present embodiment, the SubPlayItem is used only for audio post recording. The number of the SubPlayItems owned by one Virtual PlayList must be 0 or 1.

[0253] The PlayList is hereinafter explained. Fig. 25 shows the PlayList syntax. The version_number indicates four character letters indicating the version numbers of the PlayList(). The version_number is encoded to "0045" in association with ISO 646. Length is a 32-bit unsigned integer indicating the total number of byte of the PlayList() as from directly after the length field to the trailing end of the PlayList(). The PlayList_type, one example of which is shown in Fig. 26, is an 8-bit field indicating the PlayList type.

[0254] The CPI_type is a one-bit flag indicating the value of the CPI_type of the Clip referenced by the PlayItem() and by the SubPlayItem(). The CPI_types defined in the CPIs of all Clips referenced by one PlayList must be of the same values. The number_of_PlayItems is a 16-bit field indicating the number of PlayItems present in the PlayList.

[0255] The PlayItem_id corresponding to the preset PlayItem() is defined by the sequence in which the PlayItem() appears in the for-loop containing the PlayItem(). The PlayItem_id begins with 0. The number_of_SubPlayItems is a 16-bit field indicating the number of SubPlayItems in the PlayList. This value is 0 or 1. An additional audio stream path (audio stream path) is a sort of a sub path.

[0256] The UIAppInfoPlayList of the PlayList syntax shown in Fig. 25 is explained. The UIAppInfoPlayList stores parameters of the user interface application concerning the PlayList. Fig. 27

shows the syntax of the UIAppInfoPlayList. The character_set is an 8-bit field indicating the method for encoding character letters encoded in the PlayList_name field. The encoding method corresponds to the values conforming to the table shown in Fig. 19.

[0257] The name_length is an 8-bit field indicating the byte length of the PlayList name indicated in the PlayList_name field. The PlayList_name field shows the appellation of the PlayList. The number of bytes of the number of the name_length counted from left of the field is the number of valid characters and indicates the PlayList appellation. The values next following these valid character letters may be any values.

[0258] The record_time_and_date is a 56-bit field storing the date and time on which the PlayList was recorded. This field is 14 numerical figures for year/month/day/hour/minute/second encoded in binary coded decimal (BCD). For example, 2001/12/23:01:02:03 is encoded to "0x20011223010203".

[0259] The duration is a 24-bit field indicating the total replay time of the PlayList in terms of hour/minute/second as a unit. This field is six numerical figures encoded in binary coded decimal (BCD). For example, 01:45:30 is encoded to "0x014530".

[0260] The valid_period is a 32-bit field indicating the valid time periods of the PlayList. This field is 8 numerical figures encoded in 4-bit binary coded decimal (BCD). The valid_period is used in the recording and/or reproducing apparatus 1, e.g., when the PlayList, for which the valid period has lapsed, is to be automatically erased. For example, 2001/05/07 is encoded to "0x20010507".

[0261] The maker_ID is a 16-bit unsigned integer indicating the maker of the DVR player (recording and/or reproducing apparatus 1) which has been the last to update its PlayList. The value encoded to maker_ID is assigned to the licensor of the DVR format. The maker_code is a 16-bit unsigned integer indicating the model number of the DVR player which has been the last to update the PlayList. The value encoded to the maker_code is determined by the maker who has received the license of the DVR format.

[0262] If the flag of the playback_control_flag is set to 1, its PlayList is reproduced only when the user has successfully entered the PIN number. If this flag is set to 0, the user may view the PlayList without the necessity of inputting the PIN number.

[0263] If the write_protect_flag is set to 1, the contents of the PlayList are not erased nor changed except the write_protect_flag. If this flag is set to 0, the user is free to erase or change the PlayList. If this flag is set to 1, the recording and/or reproducing apparatus 1 demonstrates a message requesting re-confirmation by the user before the user proceeds to erase, edit or overwrite the PlayList.

[0264] The Real PlayList, in which the write_protect_flag is set to 0, may exist, the Virtual PlayList, referencing the Clip of the Real PlayList may exist, and the write_protect_flag of the Virtual PlayList may be set to 1. If the user desires to erase the Real PlayList, the recording and/or reproducing apparatus 1 issues an alarm to the user as to the presence of the aforementioned Virtual PlayList or "minimizes" the Real PlayList before erasing the Real PlayList.

[0265] If the `is_played_flag` is set to 1, as shown in Fig. 28B, it indicates that the PlayList was reproduced at least once since it was recorded, whereas, if it is set to 0, it indicates that the PlayList was not reproduced even once since it was recorded.

[0266] Archive is a two-bit field indicating whether the PlayList is an original or a copy, as shown in Fig. 28C. The field of `ref_thumbnail_index` indicates the information of a thumbnail picture representative of the PlayList. If the `ref_thumbnail_index` field is of a value other than 0xFFFF, a thumbnail picture representative of the PlayList is added in the PlayList, with the PlayList being stored in the `menu.thmb` file. The picture is referenced using the value of `ref_thumbnail_index` in the `menu.thmb` file. If the `ref_thumbnail_index` field is 0xFFFF, no thumbnail picture representative of the PlayList is added in the PlayList.

[0267] The PlayItem is hereinafter explained. One `PlayItem()` basically contains the following data: `Clip_Information_file_name` for specifying the filename of the Clip, IN-time and OUT-time, paired together to specify the playback domain of Clip, `STC_sequence_id` referenced by IN-time and OUT-time in case the `CPI_type` defined in `PlayList()` is `EP_map` type, and `Connection_Condition` indicating the connection condition of previous PlayItem and current PlayItem.

[0268] If PlayList is made up of two or more PlayItems, these PlayItems are arrayed in a row, without temporal gap or overlap, on the global time axis of the PlayList. If `CPI_type` defined in the PlayList is `EP_map` type and the current PlayList does not have the `BridgeSequence()`, the IN-time and OUT-time pair must

indicate the same time on the STC continuous domain as that specified by the STC_sequence_id. Such instance is shown in Fig. 29.

[0269] Fig. 30 shows such a case in which the CPI_type defined by PlayList() and, if the current PlayItem has the BridgeSequence(), the rules as now explained are applied. The IN_time of the PlayItem previous to the current PlayItem, shown as IN_time1, indicates the time in Bridge-Clip specified in the BridgeSequenceInfo() of the current PlayItem. This OUT_time must obey the encoding limitations which will be explained subsequently.

[0270] The IN_time of the current PlayItem, shown as IN_time2, indicates the time in Bridge-Clip specified in the BridgeSequenceInfo() of the current PlayItem. This IN_time also must obey the encoding limitations as later explained. The OUT_time of PlayItem of the current PlayItem, shown as OUT_time2, indicates the time on the STC continuous domain specified by STC_sequenbce_id of the current PlayItem.

[0271] If the CPI_type of PlayList() is TU_map type, the IN_time and OUT_time of PlayItem, paired together, indicate the time on the same Clip AV stream, as shown in Fig. 31.

[0272] The PlayItem syntax is as shown in Fig. 32. As to the syntax of the PlayItem, the field of the Clip_information_file_name indicates the filename of the Clip Information. The Clip_stream_type defined by the ClipInfo() of this Clip Information file must indicate the Clip AV stream.

[0273] The STC_sequence_id is an 8-bit field and indicates the STC_sequence_id of the continuous STC domain referenced by the PlayItem. If the CPI_type specified in the PlayList() is TU_map

type, this 8-bit field has no meaning and is set to 0. `IN_time` is a 32-bit field and used to store the playback start time of `PlayItem`. The semantics of `IN_time` differs with `CPI_type` defined in the `PlayList()`, as shown in Fig. 33.

[0274] OUT_time is a 32-bit field and is used to store the playback end time of PlayItem. The semantics of OUT_time differs with CPI_type defined in the PlayList(), as shown in Fig. 34.

[0275] Connection_condition is a 2-bit field indicating the connection condition between the previous PlayItem and the current PlayItem, as shown in Fig. 35. Figs. 36A to 36D illustrate various states of Connection_condition shown in Fig. 35.

[0276] BridgeSequenceInfo is explained with reference to Fig. 37. This BridgeSequenceInfo is the ancillary information of the current PlayItem and includes the following information. That is, BridgeSequenceInfo includes Bridge_Clip_Information_file_name for specifying the Bridge_Clip_AV_file and a Bridge_Clip_Information_file_name specifying the corresponding Clip Information file (Fig. 45).

[0277] It is also an address of a source packet on the Clip AV stream referenced by the previous PlayItem. Next to this source packet is connected the first source packet of the Bridge-Clip AV stream. This address is termed the RSPN_exit_from_previous_Clip. It is also an address of the source packet on the Clip AV stream referenced by the current PlayItem. Ahead of this source packet is connected the last source packet of the Bridge_clip AV stream file. This address is termed RSPN_enter_to_current_Clip.

[0278] In Fig. 37, RSPN_arrival_time_discontinuity indicates an address of a source packet in the Bridge Clip AV stream where

there is a discontinuous point in the arrival time base. This address is defined in the ClipInfo() (Fig. 46).

[0279] Fig. 38 shows the syntax of the BridgeSequenceInfo. The field of Bridge_Clip_Information_file_name indicates the filename of the Clip Information file corresponding to the Bridge_Clip_Information_file. The Clip_stream_type defined in ClipInfo() of this Clip information file must indicate 'Bridge_Clip AV stream'.

[0280] The 32-bit field of the RSPN_exit_from_previous_Clip is a relative address of a source packet on the Clip AV stream referenced by the previous PlayItem. Next to this source packet is connected the first source packet of the Bridge_Clip AV stream file. The RSPN_exit_from_previous_Clip has a size based on the source packet number as a unit, and is counted with the value of the offset_SPN defined in the ClipInfo() from the first source packet of the Clip AV stream file referenced by the previous PlayItem.

[0281] The 32-bit field of RSPN_enter_to_curent_Clip is the relative address of the source packet on the Clip AV stream referenced by the current PlayItem. Ahead of this source packet is connected the last source packet of the Bridge_Clip_AV stream file. The RSPN_enter_to_curent_Clip has a size that is based on the source packet number as a unit. The RSPN_enter_to_curent_Clip is counted with the value of the offset_SPN, defined in the ClipInfo() from the first source packet of the Clip AV stream file referenced by the current PlayItem, as an initial value.

[0282] The SubPlayItem is explained with reference to Fig. 39. The use of SubPlayItem() is permitted only if the CPI_type of the

PlayList() is the EP_map type. In the present embodiment, SubPlayItem is used only for audio post recording. The SubPlayItem() includes the following data. First, it includes Clip_Information_file_name for specifying the Clip referenced by the sub path in the PlayList.

[0283] It also includes SubPath_IN_time and SubPath_OUT_time for specifying the sub path playback domain in the Clip. Additionally, it includes sync_PlayItem_id and start PTS_of_PlayItem for specifying the time of starting the sub path reproduction on the main path time axis. The Clip AV stream, referenced by the sub path, must not contain STC discontinuous points (discontinuous points of the system time base). The clocks of audio samples of the Clip used in the sub path are locked to the clocks of the audio samples of the main path.

[0284] Fig. 40 shows the syntax of the SubPlayItem. The field of the Clip_Information_file_name indicates the filename of the Clip Information file and is used by a sub path in the PlayList. The Clip_stream_type defined in this ClipInfo() must indicate the Clip AV stream.

[0285] An 8-bit field of sync_PlayItem_id indicates the sub path type. Here, only '0x00' is set, as shown in Fig. 41, while other values are reserved for future use.

[0286] The 8-bit field of sync_PlayItem_id indicates the PlayItem_id of the PlayItem containing the time of playback start of the sub path on the time axis of the main path. The value of PlayItem_id corresponding to the preset PlayItem is defined in the PlayList() (Fig. 25).

[0287] A 32-bit field of sync_start PTS_of_PlayItem denotes

the time of playback start of the sub path on the time axis of the main path, and denotes the upper 32 bits of the PTS (presentation time stamp) on the PlayItem referenced by the sync_PlayItem_id. The upper 32 bit field of the SubPath_IN_time stores the playback start time of the sub path. SubPath_IN_time denotes upper 32 bits of the PTS of 33 bits corresponding to the first presentation unit in the sub path.

[0288] The upper 32 bit field of subPath_OUT_time stores the playback end time of the sub path. SubPath_OUT_time indicates upper 32 bits of the value of the Presentation_end_TS calculated by the following equation:

$$\text{Presentation_end_TS} = \text{PTS_OUT} + \text{AU_duration}$$

where PTS_out is the PTS of the 33 bit length corresponding to the last presentation unit of the SubPath and AU_duration is the 90 kHz based display period of the last presentation unit of the SubPath.

[0289] Next, PlayListMark() in the syntax of xxxx.rpls and yyyy.vpls shown in Fig. 23 is explained. The mark information pertinent to the PlayList is stored in this PlayListMark. Turning to the syntax of the PlayListMark, shown in Fig. 42, version_number is four character letters indicating the version number of this PlayListMark(). The version_number must be encoded to "0045" in accordance with ISO 646.

[0290] Length is an unsigned 32-bit integer indicating the number of bytes of PlayListMark() from directly after the length field to the trailing end of the PlayListMark(). The number_of_PlayListMarks is a 16-bit unsigned integer indicating the number of marks stored in the PlayListMark. The number_of_PlayListMarks may be zero. The mark_type is an 8-bit

field indicating the mark type and is encoded in the table shown in Fig. 43.

[0291] A 32-bit field of mark_time_stamp stores a time stamp indicating the point specified by the mark. The semantics of the mark_time_stamp differ with CPI_type defined in the PlayList(), as shown in Fig. 44. The PlayItem_id is an 8-bit field specifying the PlayItem where the mark is put. The values of PlayItem_id corresponding to a preset PlayItem are defined in the PlayList() (see Fig. 25).

[0292] An 8-bit field of character_set shows the encoding method of character letters encoded in the mark_name field. The encoding method corresponds to values shown in Fig. 19. The 8-bit field of name_length indicates the byte length of the mark name shown in the mark_name field. The mark_name field denotes the mark name indicated in the mark_name field. The number of bytes corresponding to the number of name_lengths from the left of this field is the effective character letters and denotes the mark name. In the mark_name field, the value next following these effective character letters may be arbitrary.

[0293] The field of the ref_thumbnail_index denotes the information of the thumbnail picture added to the mark. If the field of the ref_thumbnail_index is not 0xFFFF, a thumbnail picture is added to its mark, with the thumbnail picture being stored in the mark.thmb file. This picture is referenced in the mark.thmb file, using the value of ref_thumbnail_index, as explained subsequently. If the ref_thumbnail_index field is 0xFFFF, it indicates that no thumbnail picture is added to the mark.

[0294] The Clip Information file is now explained. The

zzzzz.clpi (Clip Information file) is made up of six objects, as shown in Fig. 45. These are ClipInfo(), STC_Info(), Program(), CPI(), ClipMark() and MakersPrivateData(). For the AV stream (Clip AV stream or Bridge-Clip AV stream) and the corresponding Clip Information file, the same string of numerals "zzzzz" is used.

[0295] The syntax of zzzzz.clpi (Clip Information file), shown in Fig. 45, is now explained. The ClipInfo_Start_address indicates the leading end address of ClipInfo() with the relative number of bytes from the leading end byte of the zzzzz.clpi file as a unit. The relative number of bytes is counted from zero.

[0296] The STC_Info_Start_address indicates the leading end address of STC_Info with the relative number of bytes from the leading end byte of the zzzzz.clpi file as a unit. The ProgramInfo_Start_address indicates the leading end address of ProgramInfo() with the relative number of bytes from the leading end byte of the zzzzz.clpi file as a unit. The relative number of bytes is counted from 0. The CPI_Start_address indicates the leading end address of CPI() with the relative number of bytes from the leading end byte of the zzzzz.clpi file as a unit. The relative number of bytes is counted from zero.

[0297] The ClipMark_Start_address indicates the leading end address of ClipMark() with the relative number of bytes from the leading end byte of the zzzzz.clpi file as a unit. The relative number of bytes is counted from zero. The MakersPrivateData Start_address indicates the leading end address of MakersPrivateData() with the relative number of bytes from the leading end byte of the zzzzz.clpi file as a unit. The relative number of bytes is counted from zero. The padding_word is

inserted in accordance with the syntax of the zzzzz.clpi file. N1, N2, N3, N4 and N5 must be zero or optional positive integers. The respective padding words may also assume optional values.

[0298] The ClipInfo is now explained. Fig. 46 shows the syntax of ClipInfo. In the ClipInfo() is stored the attribute information of corresponding AV stream files (Clip AV stream or Bridge-Clip AV stream file).

[0299] Turning to the syntax of the ClipInfo shown in Fig. 46, version_number is the four character letters indicating the version number of this ClipInfo(). The version_number must be encoded to "0045" in accordance with the ISO 646. Length is a 32-bit unsigned integer indicating the number of bytes of ClipInfo() from directly at back of the length field to the trailing end of the ClipInfo(). An 8-bit field of Clip_stream_type indicates the type of the AV stream corresponding to the Clip Information file, as shown in Fig. 47. The stream types of the respective AV streams will be explained subsequently.

[0300] The 32-bit field of offset_SPN gives an offset value of the source packet number of the first source packet of the AV stream (Clip AV stream or the Bridge-Clip AV stream). When the AV stream file is first recorded on the disc, this offset_SPN must be zero.

[0301] Referring to Fig. 48, when the beginning portion of the AV stream file is erased by editing, offset_SPN may assume a value other than 0. In the present embodiment, the relative source packet number (relative address) referencing the offset_SPN is frequently described in the form of RSPNxxx, where xxx is modified such that RSPN_xxx is RSPN_EP_start. The

relative source packet number is sized with the source packet number as a unit and is counted from the first source packet number of the AV stream file with the value of the offset_SPN as the initial value.

[0302] The number of source packets from the first source packet of the AV stream file to the source packet referenced by the relative source packet number (SPN_xxx) is calculated by the following equation:

SPN_xxx = RSPN_xxx - offset SPN.

[0303] Fig. 48 shows an instance in which offset SPN is 4.

[0304] TS_recording_rate is a 24-bit unsigned integer, which affords an input/output bit rate required for the AV stream to the DVR drive (write unit 22) or from the DVR drive (readout unit 28). The record_time_and_date is a 56-bit field for storing the date and time of recording of the AV stream corresponding to the Clip and is an encoded representation of year/month/day/hour/minute in 4-bit binary coded decimal (BCD) for 14 numerical figures. For example, 2001/2/23:01:02:03 is encoded to "0x20011223010203".

[0305] The duration is a 24-bit field indicating the total playback time of the Clip by hour/minute/second based on arrival time clocks. This field is six numerical figures encoded in 4-bit binary coded decimal (BCD). For example, 01:45:30 is encoded to "0x014530".

[0306] A flag `time_controlled_flag` indicates the recording mode of an AV stream file. If this `time_controlled_flag` is 1, it indicates that the recording mode is such a mode in which the file size is proportionate to the time elapsed since recording, such that the condition is shown by the following equation:

$Ts_average_rate * 192 / 188 * (t - start_time) - \alpha \leq size_clip(t)$
 $\leq Ts_average_rate * 192 / 188 * (t - start_time) + \alpha$

where `TS_average_rate` is an average bit rate of the transport stream of the AV stream file expressed by bytes/second.

[0307] In the above equation, `t` denotes the time in seconds, while `start_time` is the time point when the first source packet of the AV stream file was recorded. The `size_clip(t)` is $10 * 192$ bytes and α is a constant dependent on `TS_average_rate`.

[0308] If `time_controlled_flag` is set to 0, it indicates that the recording mode is not controlling so that the time lapse of recording is proportionate to the file size of the AV stream. For example, the input transport stream is recorded in a transparent fashion.

[0309] If `time_controlled_flag` is set to 1, the 24-bit field of `TS_average_rate` indicates the value of `TS_average_rate` used in the above equation. If `time_controlled_flag` is set to 0, this field has no meaning and must be set to 0. For example, the variable bit rate transport stream is encoded by the following sequence: first, the transport rate is set to the value of `TS_recording_rate`. The video stream is encoded with a variable bit rate. The transport packet is intermittently encoded by not employing null packets.

[0310] The 32-bit field of `RSPN_arrival_time_discontinuity` is a relative address of a site where arrival timebase discontinuities are produced on the Bridge-Clip AV stream file. The `RSPN_arrival_time_discontinuity` is sized with the source packet number as a unit and is counted with the value of `offset_SPN` defined in the `ClipInfo()` as from the first source packet of the Bridge-Clip AV stream file. An absolute address in

the Bridge-Clip AV stream file is calculated based on the aforementioned equation:

SPN_xxx = RSPN_xxx - offset_SPN.

[0311] The 144-bit field of reserved_for_system_use is reserved for a system. If is_format_identifier_valid flag is 1, it indicates that the field of format_identifier is effective. If is_format_identifier_valid flag is 1, it indicates that the format_identifier field is valid. If

is_original_network_ID_valid flag is 1, it indicates that the field of original_network_ID is valid. If the flag is_transport_stream_ID_valid is 1, it indicates that the transport_stream_ID field is valid. If is_service_ID_valid flag is 1, it indicates that the service_ID field is valid.

[0312] If is_country_code_valid flag is 1, it indicates that the field country_code is valid. The 32-bit field of format_identifier indicates the value of format_identifier owned by a registration descriptor (defined in ISO/IEC13818-1) in the transport stream. The 16-bit field of original_network_ID indicates the value of the original_network_ID defined in the transport stream.

[0313] The 16-bit field in service_ID denotes the value of service_ID defined in the transport stream. The 24-bit field of country_code shows a country code defined by ISO3166. Each character code is encoded by ISO8859-1. For example, Japan is represented as "JPN" and is encoded to "0x4A 0x50 0x4E". The stream_format_name is 15 character codes of ISO-646 showing the name of a format organization affording stream definitions of transport streams. An invalid byte in this field has a value of '0xFF'.

[0314] Format_identifier, original_network_ID, transport_stream_ID, service_ID, country_code and stream_format_name indicate service providers of transport streams. This allows to recognize encoding limitations on audio or video streams and stream definitions of private data streams other than audio/video streams or SI (service information). This information can be used to check if the decoder is able to decode the stream. If such decoding is possible, the information may be used to initialize the decoder system before starting the decoding.

[0315] STC_Info is now explained. The time domain in the MPEG-2 transport stream not containing STC discontinuous points (discontinuous points of the system time base) is termed the STC_sequence. In the Clip, STC_sequence is specified by the value of STC_sequence_id. Figs. 50A and 50B illustrate a continuous STC domain. The same STC values never appear in the same STC_sequence, although the maximum time length of Clip is limited, as explained subsequently. Therefore, the same PTS values also never appear in the same STC_sequence. If the AV stream contains N STC discontinuous points, where $N > 0$, the Clip system time base is split into $(N+1)$ STC sequences.

[0316] STC_Info stores the address of the site where STC discontinuities (system timebase discontinuities) are produced. As explained with reference to Fig. 51, the RSPN_STC_start indicates the address and begins at a time point of arrival of the source packet referenced by the $(k+1)$ st RSPN_STC_start and ends at a time point of arrival of the last source packet.

[0317] Turning to the syntax of STC_Info shown in Fig. 52, version number is four character letters indicating the version

number of STC_Info(). The version_number must be encoded to "0045" in accordance with ISO 646.

[0318] Length is a 32-bit unsigned integer indicating the number of bytes of STC_Info() from directly after this length field to the end of STC_Info(). If CPI_type of CPI() indicates TU_map type, 0 may be set in this length field. If CPI_type of CPI() indicates EP_map type, the num_of_STC_sequences must be of a value not less than 1.

[0319] An 8-bit unsigned integer of num_of_STC_sequences indicates the number of sequences in the Clip. This value indicates the number of the for-loops next following the field. The STC_sequence_id corresponding to the preset STC_sequence is defined by the order in which appears the RSPN_STC_start corresponding to the STC_sequence in the for-loop containing the RSPN_STC_start. The STC_sequence_id commences at 0.

[0320] The 32-bit field of RSPN_STC_start indicates an address at which the STC_sequence commences on the AV stream file. RSPN_STC_start denotes an address where system time base discontinuities are produced in the AV stream file. The RSPN_STC_start may also be a relative address of the source packet having the first PCR of the new system time base in the AV stream. The RSPN_STC_start is of a size based on the source packet number and is counted from the first source packet of the AV stream file with the offset_SPN value defined in ClipInfo() as an initial value. In this AV stream file, the absolute address is calculated by the above-mentioned equation, that is

$$\text{SPN_xxx} = \text{RSPN_xxx} - \text{offset_SPN}.$$

[0321] ProgramInfo in the syntax of zzzz.clip shown in Fig. 45 is now explained with reference to Fig. 53. The time domain

having the following features in the Clip is termed program_sequence. These feature are that the value of PCR_PID is not changed, the number of video elementary streams is also not changed, the PID values in the respective video streams are not changed, the encoding information which is defined by VideoCodingInfo thereof is not changed, the number of audio elementary streams is also not changed, the PID values of the respective audio streams are not changed, and that the encoding information, which is defined by AudioCodingInfo thereof, is not changed.

[0322] Program_sequence has only one system time base at the same time point. Program_sequence has a sole PMT at the same time point. ProgramInfo() stores the address of the site where the program_sequence commences. RSPN_program_sequence_start indicates the address.

[0323] Fig. 54 illustrates the syntax of ProgramInfo. Turning to the ProgramInfo shown in Fig. 54, version_number is four character letters indicating the version number of ProgramInfo(). The version_number must be encoded to "0045" in accordance with ISO 646.

[0324] Length is a 32-bit unsigned integer indicating the number of bytes of ProgramInfo() from directly at back of this length field to the end of ProgramInfo(). If CPI_type of CPI() indicates the TU_map type, this length field may be set to 0. If the CPI_type of CPI() indicates EP_map type, the number_of_programs must be of a value not less than 1.

[0325] An 8-bit unsigned integer of number_of_program_sequences denotes the number of program_sequences in the Clip. This value indicates the number

of for-loops next following this field. If program_sequence in the Clip is not changed, 1 must be set in the number of program_sequences. A 32-bit field of RSPN_program_sequence_start is a relative address where the program sequence commences on the AV stream.

[0326] RSPN_program_sequence_start is sized with the source packet number as a unit and is counted with the value of offset_SPN defined in the ClipInfo() as from the first source packet of the AV stream file. An absolute address in the AV stream file is calculated by:

$$\text{SPN_xxx} = \text{RSPN_xxx} - \text{offset_SPN}.$$

[0327] The values of RSPN_program_sequence_start in the for-loop syntax must appear in the rising order.

[0328] A 16-bit field of PCR_PID denotes the PID of the transport packet containing an effective PCR field effective for the program_sequence. An 8-bit field of number_of_audios indicates the number of for-loops containing audio_stream_PID and AudioCodingInfo(). A 16-bit field of video_stream_PID indicates the PID of the transport packet containing a video stream effective for its program_sequence. VideoCodingInfo(), next following this field, must explain the contents of the video stream referenced by its video_stream_PID.

[0329] A 16-bit field of audio_stream_PID indicates the PID of a transport packet containing the audio stream effective for its program_sequence. The AudioCodingInfo(), next following this field, must explain the contents of the video stream referenced by its audio_stream_PID.

[0330] The order in which the values of video_stream_PID in the for-loop of the syntax must be equal to the sequence of PID

encoding of the video stream in the PMT effective for the program_sequence. Additionally, the order in which the values of audio_stream_PID appears in the for-loop of the syntax must be equal to the sequence of PID encoding of the audio stream in the PMT effective for the program_sequence.

[0331] Fig. 55 shows the syntax of VideoCodingInfo in the syntax of the ProgramInfo shown in Fig. 54. Turning to the syntax of the VideoCodingInfo shown in Fig. 55, an 8-bit field of video_format indicates the video format corresponding to video_stream_PID in ProgramInfo(), as shown in Fig. 56.

[0332] Referring to Fig. 57, an 8-bit field of frame_rate indicates the video frame rate corresponding to the video_stream_PID in ProgramInfo(). An 8-bit field of display_aspect_ratio indicates a video display aspect ratio corresponding to video_stream_PID in ProgramInfo().

[0333] Fig. 59 shows the syntax of AudioCodingInfo in the syntax of ProgramInfo shown in Fig. 54. Turning to the syntax of the AudioCoding Info shown in Fig. 59, an 8-bit field of audio_format indicates the audio encoding method corresponding to audio_stream_PID in ProgramInfo(), as shown in Fig. 60.

[0334] An 8-bit field of audio_component_type indicates an audio component type corresponding to audio_stream_PID in ProgramInfo(), as shown in Fig. 61, whilst an 8-bit field of sampling_frequency indicates an audio sampling frequency corresponding to audio_stream_PID in ProgramInfo(), as shown in Fig. 62.

[0335] The CPI (Characteristics Point Information) in the syntax of zzzzz.clip shown in Fig. 45 is explained. The CPI is used for correlating the time information in the AV stream with

the address in its file. The CPI is of two types, namely EP_map and TU_map. In Fig. 63, if CPI_type in CPI() is EP_map, its CPI() contains EP_map. In Fig. 64, if CPI_type in CPI() is TU_map, its CPI() contains TU_map. One AV stream has one EP_map or one TU_map. If the AV stream is an SESF transport stream, the corresponding Clip must own an EP map.

[0336] Turning to the syntax of CPI shown in Fig. 65, the version_number is four character letters indicating the version number of this CPI(). The version_number must be encoded to "0045" in accordance with ISO 646. Length is a 32-bit unsigned integer indicating the number of bytes as from directly after this length field to the trailing end of the CPI(). The CPI_type is a 1-bit flag and indicates the CPI type of Clip, as shown in Fig. 66.

[0337] The EP_map in the CPI syntax shown in Fig. 65 is explained. There are two types of the EP_map, that is EP_map for a video stream and an EP_map for an audio stream. The EP_map_type in the EP_map differentiates between these EP_map types. If the Clip contains one or more video streams, the EP_map for the video stream must be used. If the Clip does not contain a video stream but contains one or more audio streams, the EP_map for the audio stream must be used.

[0338] The EP_map for a video stream is explained with reference to Fig. 67. The EP_map for the video stream has data stream_PID, PTS_EP_start and RSPN_EP_start. The stream_PID shows the PID of the transport packet transmitting a video stream. The PTS_EP_start indicates the PTS of an access unit beginning from the sequence header of the video stream. The RSPN_EP_start indicates the address of a source packet including the first byte

of the access unit referenced by the PTS_EP_start in the AV stream.

[0339] A sub table, termed EP_map_for_one_stream_PID() is created from one video stream transmitted by the transport packet having the same PID to another. If plural video streams exist in the Clip, the EP_map may contain plural EP_map_for_one_stream_PID().

[0340] The EP_map for audio stream has data stream_PID, PTS_EP_start and RSPN_EP_start. The stream_PID shows a PID of a transport packet transmitting an audio stream. The PTS_EP_start shows the PTS of an accessing unit in the audio stream. The RSPN_EP_start indicates an address of a source packet containing a first byte of the access unit referenced by PTS_EP_start of the AV stream.

[0341] The sub table termed EP_map_for_one_stream_PID() is created from one audio stream transmitted by the transport packet having the same PID to another. If there exist plural audio streams in the Clip, EP_map may contain plural EP_map_for_one_stream_PID().

[0342] Turning to the relation between EP_map and STC_Info, one EP_map_for_one_stream_PID() is created in one table irrespective of discontinuous points in the STC. Comparison of the value of the RSPN_EP_start to the value of RSPN_STC_start defined in STC_Info() reveals the boundary of data of EP_map belonging to respective STC_sequences (see Fig. 68). The EP_map must have one EP_map_for_one_stream_PID() for a continuous stream range transmitted by the same PID. In the case shown in Fig. 69, program#1 and program#3 have the same video PID, however, the data range is not continuous, so that EP_map_for_one_stream_PID()

must be provided for each program.

[0343] Fig. 70 shows the EP_map syntax. By way of explanation of the EP_map syntax shown in Fig. 70, the EP_type is a 4-bit field and shows the EP_map entry point type, as shown in Fig. 71. The EP_type shows the semantics of the data field next following this field. If Clip includes one or more video stream, the EP_type must be set to 0 ('video'). Alternatively, if the Clip contains no video stream but contains one or more audio stream, then EP_type must be set to 1 ('audio').

[0344] The 16-bit field of number_of_stream_PIDs indicates the number of times of loops of the for-loop having number_of_stream_PIDs in the EP_map() as a variable. The 16-bit field of stream_PID(k) indicates the PID of the transport packet transmitting the number k elementary stream (video or audio stream) referenced by EP_map_for_one_stream_PID (num_EP_entries(k)). If EP_type is 0 ('video'), its elementary stream must be a video stream. If EP_type is equal to 1 ('audio'), its elementary stream must be the audio stream.

[0345] The 16-bit field of num_EP_entries(k) indicates the num_EP_entries(k) referenced by EP_map_entries(k). The EP_map_for_one_stream_PID_Start_address(k): this 32-bit field indicates the relative address position at which the EP_map_for_one_stream_PID(num_EP_entries(k)) begins in the EP_map(). This value is indicated by the size as from the first byte of the EP_map().

[0346] Padding_word must be inserted in accordance with the EP_map() syntax. X and Y must be optional positive integers. The respective padding words may assume any optional values.

[0347] Fig. 72 shows the syntax of EP_map_for_one_stream_PID.

By way of explanation of the syntax of the EP_map_for_one_stream_PID shown in Fig. 72, the semantics of the 32-bit field of PTS_EP_start differs with the EP_type defined by EP_map(). If EP_type is equal to 0 ('video'), this field has upper 32 bits of the 33-bit precision PTS of the access unit beginning with a sequence header of the video stream. If the EP_type is equal to 1 ('audio'), this field has upper 32 bits of PTS of 33 bit precision of the access unit of the audio stream.

[0348] The semantics of the 32-bit field of RSPN_EP_start differs with the EP_type defined in EP_map(). If EP_type is equal to 0 ('video'), this field indicates the relative address of the source packet including the first byte of the sequence header of the access unit referenced by the PTS_EP_start in the AV stream. Alternatively, if EP_type is equal to 1 ('audio'), this field indicates the relative address of the source packet containing the first byte in the audio stream of the access unit referenced by the PTS_EP_start in the AV stream.

[0349] RSPN_EP_start is of a size which is based on the source packet number as a unit, and is counted from the first source packet of the AV stream file, with the value of the offset_SPN, defined in ClipInfo(), as an initial value. The absolute address in the AV stream file is calculated by

SPN_xxx = RSPN_xxx - offset_SPN.

[0350] It is noted that the value of the RSPN_EP_start in the syntax must appear in the rising order.

[0351] The TU_map is now explained with reference to Fig. 73. TU_map forms a time axis based on the source packet arrival time clock (timepiece of the arrive time base). This time axis is termed TU_map_time_axis. The point of origin of TU_map_time_axis

is indicated by `offset_time` in the `TU_map()`. `TU_map_time_axis` is divided in a preset unit as from `offset_time`, this unit being termed `time_unit`.

[0352] In each `time_unit` in the AV stream, addresses on the AV stream file of the source packet in the first complete form are stored in `TU_map`. These addresses are termed `RSNP_time_unit_start`. The time at which begins the $k (k \geq 0)$ th `time_unit` on the `TU_map_time_axis` is termed `TU_start_time(k)`. This value is calculated based on the following equation:

$$\text{TU_start_time}(k) = \text{offset_time} + k * \text{time_unit_size}.$$

[0353] It is noted that `TU_start_time(k)` has a precision of 45 kHz.

[0354] Fig. 74 shows the syntax of `TU_map`. By way of explanation of the `TU_map` syntax shown in Fig. 74, the 32-bit field of `offset_time` gives an offset time relative to `TU_map_time_axis`. This value indicates the offset time relative to the first `time_unit` in the Clip. The `offset_time` is of a size based on 45 kHz clock derived from the 27 MHz precision arrival time clocks as a unit. If the AV stream is to be recorded as new Clip, `offset_time` must be set to 0.

[0355] The 32-bit field of `time_unit_size` affords the size of the `time_unit`, and is based on 45 kHz clocks, derived from the 27 MHz precision arrival time clocks, as a unit. Preferably, `time_unit_size` is not longer than one second (`time_unit_size \leq 45000`). The 32 bit field of `number_of_time_unit_entries` indicates the number of entries stored in `TU_map()`.

[0356] The 32-bit field of `RSNP_time_unit_start` indicates the relative address of a site in the AV stream at which begins each `time_unit`. `RSNP_time_unit_start` is of a size based on the source

packet number as unit and is counted with the value of offset_SPN defined in ClipInfo() as from the first source packet of the AV stream file as an initial value. The absolute address in the AV stream file is calculated by

SPN_xxx = RSPN_xxx - offset_SPN.

[0357] It is noted that the value of RSN_time_unit_start in the for-loop of the syntax must appear in the rising order. If there is no source packet in the number (k+1) time_unit, the number (k+1) RSN_time_unit_start must be equal to the number k RSPN_time_unit_start.

[0358] By way of explanation of the ClipMark in the syntax of zzzzz.clip shown in Fig. 45, the ClipMark is the mark information pertinent to Clip and is stored in the ClipMark. This mark is not set by a user, but is set by a recorder (recording and/or reproducing apparatus 1).

[0359] Fig. 75 shows the ClipMark syntax. By way of explanation of the ClipMark syntax shown in Fig. 75, the version_number is four character letters indicating the version number of this ClipMark. The version_number must be encoded in accordance with ISO 646 to "0045".

[0360] Length is a 32-bit unsigned integer indicating the number of bytes of the ClipMark() as from directly after the length field to the trailing end of ClipMark(). The number_of_Clip_marks is a 16-bit unsigned integer indicating the number of marks stored in ClipMark and may be equal to 0. Mark_type is an 8-bit field indicating the mark type and is encoded in accordance with the table shown in Fig. 76.

[0361] Mark_time_stamp is a 32-bit field and stores the time stamp indicating a pointer having a specified mark. The

semantics of mark_time_stamp differs with CPI_type in the PlayList(), as shown in Fig. 77.

[0362] If CPI_type in CPI() indicates the EP_map type, this 8-bit field indicates the STC_sequence_id of the continuous STC domain where there is placed mark_time_stamp. If CPI_type in CPI() indicates TU_map type, this 8-bit field has no meaning but is set to 0. The 8-bit field of character_set indicates the indicating method of character letters encoded in the mark_name field. The encoding method corresponds to the value shown in Fig. 19.

[0363] The 8-bit field of name_length indicates the byte length of the mark name shown in the mark_name field. This mark_name field indicates the mark name. The byte number corresponding to the number of the name_length from the left of this field is the effective character number and denotes the mark name. In the mark_name field, the values next following these effective character letters may be arbitrary.

[0364] The field of ref_thumbnail_index indicates the information of the thumbnail picture appended to the mark. If the ref_thumbnail_index field is of a value different from 0xFFFF, a thumbnail picture is added to its mark, with the thumbnail picture being stored in the mark.thmb file. This picture is referenced using the value of ref_thumbnail_index in the mark.thmb file. If the ref_thumbnail_index field is of a value equal to 0xFFFF, a thumbnail picture is not appended to its mark.

[0365] MakerPrivateData has already been explained with reference to Fig. 22 and hence is not explained here specifically.

[0366] Next, thumbnail_information is explained. A thumbnail picture is stored in a menu.thmb file or in a mark.thmb file. These files are of the same syntax structure and own a sole Thumbnail(). The menu.thmb file stores a picture representing respective PlayLists. The totality of menu thumbnails are stored in the sole menu.thmb file.

[0367] The mark.thmb file stores a mark thumbnail picture, that is, a picture representing a mark point. The totality of mark thumbnails corresponding to the totality of PlayLists and Clips are stored in the sole mark.thmb file. Since the

thumbnails are frequently added or deleted, the operation of addition and partial deletion must be executable readily and speedily. For this reason, Thumbnail() has a block structure. Picture data is divided into plural portions each of which is stored in one tn_block. One picture data is stored in consecutive tn_blocks. In the string of tn_blocks, there may exist a tn_block not in use. The byte length of a sole thumbnail picture is variable.

[0368] Fig. 78 shows the syntax of menu.thmb and mark.thmb and Fig. 79 the syntax of Thumbnail in the syntax of menu.thmb and mark.thmb shown in Fig. 78. By way of explanation of the syntax of Thumbnail, shown in Fig. 79, version_number is four character letters denoting the version number of this Thumbnail(). Version_number must be encoded to "0045" in accordance with ISO 646.

[0369] Length is a 32-bit unsigned integer indicating the number of bytes of MakerPrivateData() as from directly at back of the length field up to the trailing end of Thumbnail(). Tu_block_start_address is a 32-bit unsigned integer indicating the leading end byte address of the first tn_block, in terms of the relative number of bytes as from the leading end byte of Thumbnail() as a unit. The number of relative bytes is counted from 0. Number_of_thumbnails is a 16-bit unsigned integer which gives the number of entries of a thumbnail picture contained in Thumbnail().

[0370] Tu_block_size is a 16-bit unsigned integer which gives the size of one tn_block, in terms of 1024 bytes as a unit. If, for example, tn_block_size = 1, it indicates that the size of one tn_block is 1024 bytes. Number_of_tn_blocks is a 116-bit

unsigned integer indicating the number of entries of tn_block in this Thumbnail(). Thumbnail_index is a 16-bit unsigned integer indicating the index number of the thumbnail picture represented by the thumbnail information for one for-loop beginning from the thumbnail_index field. The value 0xFFFF must not be used as Thumbnail_index. This Thumbnail_index is referenced by ref_thumbnail_index in UIAppInfoVolume(), UIAppInfoPlayList(), PlayListMark() and ClipMark().

[0371] Thumbnail_picture_format is an 8-bit unsigned integer representing the picture format of a thumbnail picture and assumes a value shown in Fig. 80. In the table, DCF and PNG are allowed only in menu.thmb. The mark thumbnail must assume the value of "0x00" (MPEG-2 Video 1-picture).

[0372] Picture_data_size is a 32-bit unsigned integer indicating the byte length of a thumbnail picture in terms of bytes as a unit. Start_tn_block_number is a 16-bit unsigned integer indicating the tn_block number of the tn_block where data of the thumbnail picture begins. The leading end of the thumbnail picture data must coincide with the leading end of the tn_block. The tn_block number begins from 0 and is relevant to the value of a variable k in the for-loop of tn_block.

[0373] X_picture_length is a 16-bit unsigned integer indicating the number of pixels in the horizontal direction of a frame of a thumbnail picture. Y_picture_length is a 16-bit unsigned integer indicating the number of pixels in the vertical direction of a frame of a thumbnail picture. Tn_block is an area in which to store a thumbnail picture. All tn_block in the Thumbnail() are of the same size (fixed length) and are of a size defined by tn_block_size.

[0374] Figs. 81A and 81B schematically show how thumbnail picture data are stored in tn_block. If, as shown in Figs. 81A and 81B, the thumbnail picture begins at the leading end of tn_block, and is of a size exceeding 1 tn_block, it is stored using the next following tn_block. By so doing, data with a variable length can be managed as fixed length data, so that the editing of deletion can be coped with by simpler processing.

[0375] An AV stream file is now explained. The AV stream file is stored in the "M2TS" directory (Fig. 14). There are two types of AV stream files, namely, a Clip A stream file and a Bridge-Clip AV stream file. Both AV streams must be of the structure of DVR MPEG-2 transport stream file as hereinafter defined.

[0376] First, the DVR MPEG2 transport stream is explained. The structure of the DVR MPEG-2 transport stream is shown in Fig. 82. The AV stream file has the structure of a DVR MPEG 2 transport stream. The DVR MPEG 2 transport stream is made up of an integer number of Aligned units. The size of the Aligned unit is 6144 bytes (2048*3 bytes). The Aligned unit begins from the first byte of the source packet. The source packet is 192 bytes long. One source packet is comprised of TP_extra_header and a transport packet. TP_extra_header is 4 bytes long, with the transport packet being 188 bytes long.

[0377] One Aligned unit is made up of 32 source packets. The last Aligned unit in the DVR MPEG 2 transport stream is also made up of 32 source packets. Therefore, the DVR MPEG 2 transport stream ends at a boundary of the Aligned unit. If the number of the transport packets of the input transport stream recorded on a disc is not a multiple of 32, a source packet having a null packet (transport packet of PID = 0x1FFFF) must be used as the

last Aligned unit. The file system must not use excess information in the DVR MPEG 2 transport stream.

[0378] Fig. 83 shows a recorder model of the DVR MPEG 2 transport stream. The recorder shown in Fig. 83 is a conceptual model for prescribing the recording process. The DVR MPEG 2 transport stream obeys this model.

[0379] The input timing of the MPEG 2 transport stream is now explained. The input MPEG 2 transport stream is a full transport stream or a partial transport stream. The input MPEG 2 transport stream must obey the ISO/IEC13818-1 or ISO/IEC 13818-9. The number i byte of the MPEG 2 transport stream is input simultaneously at time $t(i)$ to T-STD (transport stream system target decoder provided for in ISO/IEC13818-1) and to the source packetizer. R_{pk} is an instantaneous maximum value of the input rate of the transport packet.

[0380] A 27 MHz PLL 52 generates a frequency of 27 MHz clock. The 27 MHz clock frequency is locked at a value of the program clock reference (PCR) of the MPEG 2 transport stream. An arrival time clock counter 53 counts the pulses of the 27 MHz frequency. $Arrival_time_clock(i)$ is a count value of the arrival time clock counter at time $t(i)$.

[0381] A source packetizer 54 appends TP_extra_header to the totality of the transport packets to create a source packet. $Arrival_time_stamp$ indicates the time when the first byte of the transport packet reaches both the T-STD and the source packetizer. $Arrival_time_stamp(k)$ is a sampled value of the $Arrival_time_clock(k)$ as represented by the following equation:
$$arrival_time_stamp(k) = arrival_time_clock(k) \% 230$$
where k denotes the first byte of the transport packet.

[0382] If the time separation between two neighboring transport packets is $230/2$ 7000000 sec (about 40 sec) or longer, the difference of the arrival_time_stamp of the two transport packets should be set to $230/2$ 7000000 sec. The recorder is provided for such case.

[0383] A smoothing buffer 55 smoothes the bit rate of the input transport stream. The smoothing buffer must not overflow. Rmax is the output bit rate of the source packet from the smoothing buffer when the smoothing buffer is not null. If the smoothing buffer is null, the output bit rate of the smoothing buffer is 0.

[0384] Next, the parameters of the recorder model of the DVR MPEG 2 transport stream are explained. The value of Rmax is given by TS_recording_rate as defined in ClipInfo() associated with the AV stream file. This value may be calculated from the following equation:

$$Rmax = TS_recording_rate * 192 / 188$$

where the value of TS_recording_rate is of a size in bytes/second.

[0385] If the input transport stream is an SESF transport stream, Rpk must be equal to TS_recording_rate as defined in ClipInfo() associated with the AV stream file. If the input transport stream is not an SESF transport stream, reference may be made to values defined, e.g., in a descriptor of the MPEG 2 transport stream, such as maximum_bit_rate_descriptor or partial_stream_descriptor for this value.

[0386] If the input transport stream is an SESF transport stream, the smoothing buffer size is 0. If the input transport stream is not an SESF transport stream, reference may be made to

values defined in the descriptor of the MPEG 2 transport stream, such as, for example, the values defined in the smoothing_buffer_descriptor, short_smoothing_buffer_descriptor or in the partial_transport_stream_descriptor.

[0387] For the recorder and the player (reproducing apparatus), a sufficient size buffer needs to be provided. The default buffer size is 1536 bytes.

[0388] Next, a player model of the DVR MPEG 2 transport stream is explained. Fig. 84 shows a player model of the DVR MPEG 2 transport stream. This is a conceptual model for prescribing the reproduction process. The DVR MPEG 2 transport stream obeys this model.

[0389] A 27 MHz crystal 61 generates the frequency of 27 MHz. An error range of the 27MHz frequency must be ± 30 ppm (2.7000000 ± 810 Hz). The arrival time clock counter 62 is a binary counter for counting the pulses of the frequency of 27 MHz. Arrival_time_clock(i) is a count value of the arrival time clock counter at time $t(i)$.

[0390] In the smoothing buffer 64, Rmax is the input bit rate of the source packet to the smoothing buffer when the smoothing buffer is not full. If the smoothing buffer is full, the input bit rate to the smoothing buffer is 0.

[0391] By way of explaining the output timing of the MPEG 2 transport stream, if the arrival_time_stamp of the current source packet is equal to 30 bits on the LSB side of arrival_time_clock(i), the transport packet of the source packet is removed from the smoothing buffer. Rpk is an instantaneous maximum value of the transport packet rate. The overflow of the smoothing buffer is not allowed.

[0392] The parameters of the player model of the DVR MPEG 2 transport stream are the same as those of the recorder model of the DVR MPEG 2 transport stream described above.

[0393] Fig. 85 shows the syntax of the source packet. `Transport_packet()` is an MPEG 2 transport stream provided in ISO/IEC 13818-1. The syntax of `TP_extra_header` in the syntax of the source packet shown in Fig. 85 is shown in Fig. 86. By way of explaining the syntax of the `TP_extra_header`, shown in Fig. 86, `copy_permission_indicator` is an integer representing the copying limitation of the payload of the transport packet. The copying limitation may be copy free, no more copy, copy once or copying prohibited. Fig. 87 shows the relation between the value of `copy_permission_indicator` and the mode it designates.

[0394] `Copy_permission_indicator` is appended to the totality of transport packets. If the input transport stream is recorded using the IEEE1394 digital interface, the value of `copy_permission_indicator` may be associated with the value of EMI (encryption mode indicator). If the input transport stream is recorded without employing the IEEE1394 digital interface, the value of `copy_permission_indicator` may be associated with the value of the CCI embedded in the transport packet. If an analog signal input is self-encoded, the value of `copy_permission_indicator` may be associated with the value of CGMS-A of the analog signal.

[0395] `Arrival_time_stamp` is an integer having a value as specified by `arrival_time_stamp` in the following equation:

$$\text{arrival_time_stamp}(k) = \text{arrival_time_clock}(k) \% 230.$$

[0396] By way of defining the ClipAV stream, the ClipAV stream must have a structure of the DVR MPEG 2 transport stream defined

as described above. `Arrival_time_clock(i)` must increase continuously in the Clip AV stream. If there exists a discontinuous point of the system time base (STC base) in the Clip AV stream, `arrival_time_clock(i)` in the Clip AV stream must increase continuously.

[0397] The maximum value of the difference of the `arrival_time_clock(i)` between the beginning and the end of the Clip AV stream must be 26 hours. This limitation guarantees that, if there is no discontinuous point in the system time base (STC base) in the MPEG 2 transport stream, the PTS (presentation time stamp) of the same value never appears in the Clip AV stream. The MPEG 2 system standard provides that the PTS has a wraparound period of 233/90000 sec (about 26.5 hours).

[0398] By way of defining the Bridge-Clip AV stream, the Bridge-Clip AV stream must have a structure of the DVR MPEG 2 transport stream defined as described above. The Bridge-Clip AV stream must include a discontinuous point of one arrival time base. The transport stream ahead and at back of the discontinuous point of the arrival time base must obey the encoding limitations and the DVR-STD as later explained. The present embodiment supports the video-audio seamless connection between PlayItems being edited. Seamless connection between PlayItems guarantees "continuous data supply" to the player/decoder and "seamless decoding processing". The "continuous data supply" is the capability of guaranteeing data supply to the decoder at a bit rate necessary to prevent buffer underflow. In order to enable data to be read out from the disc as data real-time properties are assured, data is to be stored in terms of a continuous block of a sufficiently large size as a

unit.

[0399] The "seamless decoding processing" means the capability of a player in displaying audio video data recorded on the disc without producing pause or gap in the playback output of the decoder.

[0400] The AV stream, referenced by the seamless connected PlayItems, is explained. Whether or not the seamless display of a previous PlayItem and the current PlayItem is guaranteed may be verified from the connection_condition field defined in the current PlayItem. There are two methods for seamless connection of PlayItems, that is, a method employing Bridge-Clip and a method not employing Bridge-Clip.

[0401] Fig. 88 shows the relation between the previous PlayItem and the current PlayItem in case of employing Bridge-Clip. In Fig. 88, the stream data, read out by the player, is shown shaded. In Fig. 88, TS1 is made up of shaded stream data of the Clip1 (Clip AV stream) and shaded stream data previous to RSPN_arrival_time_discontinuity.

[0402] The shaded stream data of Clip1 of TS1 is stream data from an address of a stream required for decoding the presentation unit corresponding to IN_item of the previous PlayItem (shown as IN-time1 in Fig. 88) up to the source packet referenced by RSPN_exit_from_previous_Clip. The shaded stream data prior to RSPN_arrival_time_discontinuity of Bridge-Clip contained in TS1 is stream data as from the first source packet of Bridge-Clip up to the source packet directly previous to the source packet referenced by RSPN_arrival_time_discontinuity.

[0403] In Fig. 88, TS2 is made up of shaded stream data of Clip 2 (Clip AV stream) and shaded stream data subsequent to

RSPN_arrival_time_discontinuity of Bridge-Clip. The shaded stream data as from the RSPN_arrival_time_discontinuity of Bridge-Clip contained in TS2 is stream data from the source packet referenced by RSPN_arrival_time_discontinuity to the last source packet of Bridge-Clip. The shaded stream data of Clip2 of TS2 is stream data from the source packet referenced by RSPN_enter_to_current_Clip to the address of the stream required for decoding the presentation unit corresponding to OUT_time of current PlayItem (shown by OUT_time2 in Fig. 88).

[0404] Fig. 89 shows the relation between the previous PlayItem and the current PlayItem in case of not employing Bridge-Clip. In this case, the stream data read out by the player is shown shaded. In Fig. 89, TS1 is made up of shaded stream data of the Clip1 (Clip AV stream). The shaded stream data of Clip1 of TS1 is data beginning at an address of a stream necessary in decoding a presentation unit corresponding to IN_time of the previous PlayItem, shown at IN_time1 in Fig. 89, as far as the last source packet of Clip1.

[0405] In Fig. 89, TS2 is shaded stream data of Clip2 (Clip AV stream). The shaded stream data of Clip2 of TS2 is stream data beginning at a first source packet of Clip2 as far as an address of the stream necessary for decoding the presentation unit corresponding to OUT_time of current PlayItem (shown at OUT_time2 in Fig. 89).

[0406] In Figs. 88 and 89, TS1 and T2 are continuous streams of the source packet. Next, the stream provisions of TS1 and TS2 and the connection conditions therebetween are scrutinized. First, encoding limitations for seamless connection are scrutinized. By way of limitations on the encoding structure of

a transport stream, the number of programs contained in TS1 and TS2 must be 1. The number of video streams contained in TS1 and TS2 must be 1. The number of audio streams contained in TS and TS2 must be 2 or less. The numbers of the audio streams contained in TS1 and TS2 must be equal to each other. It is also possible for elementary streams or private streams other than those depicted above to be contained in TS1 and/or TS2.

[0407] The limitations on the video bit stream are now explained. Fig. 90 shows a typical seamless connection indicated by a picture display sequence. In order for a video stream to be demonstrated seamlessly in the vicinity of a junction point, unneeded pictures displayed at back of OUT_time1 (OUT_time of Clip1) and ahead of IN_time2 (IN_time of Clip2) must be removed by a process of re-encoding the partial stream of the Clip in the vicinity of the junction point.

[0408] Fig. 91 shows an embodiment of realizing seamless connection using BridgeSequence. The video stream of Bridge-Clip previous to RSPN_arrival_time_discontinuity is comprised of an encoded video stream up to a picture corresponding to OUT_time1 of Clip1 of Fig. 90. This video stream is connected to the video stream of previous Clip1 and is re-encoded to form an elementary stream conforming to the MPEG2 standard.

[0409] The video stream of Bridge-Clip subsequent to RSPN_arrival_time_discontinuity is made up of an encoded video stream subsequent to a picture corresponding to IN_time2 of Clip2 of Fig. 90. The decoding of this video stream can be started correctly for connecting the video stream to the next following Clip2 video stream. Re-encoding is made such that a sole continuous elementary stream conforming to MPEG 2 standard will

be formed. For creating Bridge-Clip, several pictures in general need to be re-encoded, whilst other pictures can be copied from the original Clip.

[0410] Fig. 92 shows an embodiment of realizing seamless connection without employing BridgeSequence in the embodiment shown in Fig. 90. The Clip1 video stream is comprised of an encoded video stream as far as the picture corresponding to OUT_time1 of Fig. 90 and is re-encoded so as to give an elementary stream conforming to the MPEG2 standard. In similar manner, the video stream of Clip2 is made up of encoded bit streams subsequent to the picture associated with IN_time2 of Clip2 of Fig. 90. These encoding bit streams are already re-encoded to give a sole continuous elementary stream conforming to the MPEG2 standard.

[0411] By way of explaining encoding limitations of the video stream, the frame rates of the video streams of TS1 and TS2 must be equal to each other. The video stream of TS1 must be terminated at sequence_end_code. The video stream of TS2 must commence at Sequence header, GOP Header and with an I-picture. The video stream of TS2 must commence at a closed GOP.

[0412] The video presentation units defined in a bit stream (frame or field) must be continuous with a junction point in-between. No gap of the fields or frames are allowed to exist at junction points. In case of encoding employing 3-2 pulldown, it may be necessary to rewrite "top_field_first" and "repeat_first_field" flags. Alternatively, local re-encoding may be made to prevent field gaps from being produced.

[0413] By way of explaining encoding limitations on the audio bit stream, the audio sampling frequency of TS1 and that of TS2

must be equal to each other. The audio encoding method of TS1 and that of TS2 (for example, MPEG1 layer 2, AC-3, SESF LPCM and AAC) must be equal to each other.

[0414] By way of explaining encoding limitations on MPEG-2 transport stream, the last audio frame of the audio stream of TS1 must contain audio samples having a display timing equal to the display end time of the last display picture of TS1. The first audio frame of the audio stream of TS2 must contain an audio sample having a display timing equal to the display start timing of the first display picture of TS2.

[0415] At a junction point, no gap may be allowed to exist in a sequence of the audio presentation units. As shown in Fig. 93, there may be an overlap defined by the length of the audio presentation unit less than two audio frame domains. The first packet transmitting an elementary stream of TS2 must be a video packet. The transport stream at the junction point must obey the DVR-STD which will be explained subsequently.

[0416] By way of explaining limitations on the Clip and Bridge-Clip, no discontinuities in the arrival time base are allowed to exist in TS1 or in TS2.

[0417] The following limitations are applied only to the case of employing the Bridge-Clip. The Bridge-Clip AV stream has a sole discontinuous point in the arrival time base only at a junction point of the last source packet of TS1 and the first source packet of TS2. The SPN_arrival_time_discontinuity defined in ClipInfo() represents an address of the discontinuous point, which must represent the address referencing the first source packet of TS2.

[0418] The source packet referenced by

RSPN_exit_from_previous_Clip defined in BridgeSequenceInfo() may be any source packet in Clip1. It is unnecessary for this source packet to be a boundary of the Aligned unit. The source packet referenced by RSPN_enter_to_current_Clip defined in BridgeSequenceInfo() may be any source packet in Clip2. It is unnecessary for this source packet to be a boundary of the Aligned unit.

[0419] By way of explaining limitations on PlayItem, the OUT_time of the previous PlayItem (OUT_time 1 shown in Fig. 89) must represent the display end time of the last video presentation unit of TS1. The IN_time of the current PlayTime (IN_time2 shown in Fig. 88 and 89) must represent the display start time of the first presentation unit of TS2.

[0420] By way of explaining the limitations on the data allocation in case of employing Bridge-Clip by referring to Fig. 94, the seamless connection must be made to guarantee continuous data supply by the file system. This must be realized by arranging the Bridge-Clip AV stream, connecting to Clip1 (Clip AV stream file) and Clip2 (Clip AV stream file), such as to satisfy data allocation prescriptions.

[0421] RSPN_exit_from_previous_Clip must be selected so that the stream portion of Clip1 (Clip AV stream file) previous to RSPN_exit_from_previous_Clip will be arranged in a continuous area not less than half fragment. The data length of the Bridge-Clip AV stream must be selected so that the data will be arranged in the continuous area not less than half fragment. RSPN_enter_to_current_Clip must be selected so that the stream portion of Clip2 (Clip AV stream file) subsequent to RSPN_enter_to_current_Clip will be arranged in a continuous area

not less than half fragment.

[0422] By way of explaining data allocation limitations in case of seamless connection not employing Bridge-Clip, by referring to Fig. 95, the seamless connection must be made so as to guarantee continuous data supply by the file system. This must be realized by arranging the last portion of the Clip1 (Clip AV stream file) and the first portion of the Clip2 (Clip AV stream file) so that the provisions on data allocation will be met.

[0423] The last stream portion of Clip1 (Clip AV stream file) must be arranged in a continuous area not less than one half fragment. The first stream portion of Clip2 (Clip AV stream file) must be arranged in a continuous area not less than one half fragment.

[0424] In a case where digital AV signals each having a predetermined bit rate are fragmented and recorded on a disc, to assure that the recorded digital AV signals can be read out from the recording medium 100 in the predetermined bit rate, the size of one continuous recording area must satisfy the following condition:

$$S*8 / (S*8/Rud + Ts) \geq Rmax$$

where

S: minimum size of one continuous recording area [Byte]

Ts: access time of full stroke from one recording area to next recording area [second]

Rud: bit rate for reading out from recording medium [bit/second]

Rmax: bit rate of AV stream [bit/second].

[0425] That is, the data must be arranged so that S byte or

more data in the AV stream may be recorded continuously on the disc. The data must be arranged so that the size of the half fragment may be 8 byte or more.

[0426] Next, DVR-STD is explained. This DVR-STD is a conceptual model for modeling the decoding processing in the generation and verification of the DVR MPEG 2 transport stream. The DVR-STD is also a conceptual model for modeling the decoding processing in the generation and verification of the AV stream referenced by two PlayItems seamlessly connected to each other as described above.

[0427] Fig. 96 shows a DVR-STD model. The model shown in Fig. 96 includes, as a constituent element, a DVR MPEG 2 transport stream player model. The notation of n, TBn, MBn, EBn, TBsys, Bsys, Rxn, Rbxn, Rxsy, Dn, Dsys, On and Pn(k) is the same as that defined in T-STD of ISO/IEC 13818-1, wherein n is an index number of an elementary stream and TBn is a transport buffer of the elementary stream n.

[0428] MBn is a multiplexing buffer of the elementary stream n and exists only for the video stream. EBn is an elementary stream buffer of the elementary stream n and is present only for the video stream. TBsys is a main buffer in a system target decoder for the system information for a program being decoded. Rxn is a transmission rate with which data is removed from TBn. Rbxn is a transmission rate with which the PES packet payload is removed from MBn and is present only for a video stream.

[0429] Rxsy is a transmission rate with which data is removed from TBsys. Dn is a decoder of the elementary stream n. Dsys is a decoder pertinent to the system information of a program being decoded. On is a re-ordering buffer of the video stream n. Pn(k)

is a number k presentation unit of the elementary stream.

[0430] The decoding process for DVR-STD is explained. During the time a sole DVR MPEG 2 transport stream is being reproduced, the timing of inputting the transport packet to TB1, TBn or TBsys is determined by `arrival_time_stamp` of the source packet. The prescriptions for the buffering operation of TB1, MB1, EB1, TBn, TBsys and Bsys are the same as those of the T-STD provided for in ISO/IEC 13818-1, while the prescriptions for the deciding and display operations are also the same as the T-STD provided for in ISO/IEC 13818-1.

[0431] The decoding process during the time the seamlessly connected PlayLists are being reproduced is now explained. Here, the reproduction of two AV streams referenced by the seamlessly connected PlayItems is explained. In the following explanation, the reproduction of TS1 and TS2, shown for example in Fig. 88, is explained. TS1 and TS2 are a previous stream and a current stream, respectively.

[0432] Fig. 97 shows a timing chart for inputting, decoding and display of transport packets when transferring from a given AV stream (TS1) to the next AV stream seamlessly connected thereto (TS2). During transfer from a preset AV stream (TS1) to the next AV stream seamlessly connected thereto (TS2), the time axis of the arrival time base of TS2 is not the same as the time axis of the arrival time base of TS1 (indicated by ATC1 in Fig. 97).

[0433] Moreover, the time axis of the system time base of TS2 (indicated by STC2 in Fig. 97) is not the same as the time axis of the system time base of TS1 (indicated by STC1 in Fig. 97). As the video display is required to be continuous seamlessly,

however, there may be overlap in the display time of the presentation units.

[0434] The input timing to DVR-STD is explained. During the time until time T1, that is until the inputting of the last video packet to the TB1 of DVR-STD, the input timing to the buffers of TB1, TBn or TBsys of DVR-STD is determined by arrival_time_stamp of the arrival time base of TS1.

[0435] The remaining packets of TS1 must be input to buffers of TBn or to TBsys of DVR-STD at a bit rate of TS_recording_rate (TS1). The TS_recording_rate (TS1) is the value of TS_recording_rate defined in ClipInfo() corresponding to Clip1. The time the last byte of TS1 is input to the buffer is the time T2. So, during the time between time T1 and time T2, arrival time stamp of the source packet is discounted.

[0436] If $N1$ is the number of bytes of the transport packet of $TS1$ next following the last video packet of $TS1$, the time $DT1$ from time $T1$ until time $T2$ is the time necessary for $N1$ bytes to be input completely at a bit rate of $TS_recording_rate$ ($TS1$), and is calculated in accordance with the following equation:

DT1 = T2 - T1 = N1/TS recording rate.

[0437] During the time from time T1 until time T2 (TS1), both the values of Rxn and Rxsys are changed to the value of TS_recording_rate (TS1). Except for this rule, the buffering operation is the same as that of T-STD.

[0438] At time T2, the arrival time clock counter is reset to the value of arrival_time_stamp of the first source packet of TS2. The input timing to the buffer of TB1, TBn or TBsys of DVR-STD is determined by arrival_time_stamp of the source packet of TB2. Both Rxn and Rxsys are changed to values defined in T-STD.

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[0439] By way of explaining additional audio buffering and system data buffering, the audio decoder and the system decoder need to have an additional buffering amount (data amount equivalent to one second) in addition to the buffer amount defined in T-STD in order to allow input data of a domain from time T1 to time T2.

[0440] By way of explaining the video presentation timing, the display on the video presentation unit must be continuous, that is devoid of gaps, through the junction point. It is noted that STC1 is the time axis of the system time base of TS1 (indicated as STC1 in Fig. 9), while STC2 is the time axis of the system time base of TS2 (shown at STC2 in Fig. 97; correctly, STC2 begins at time the first PCR of TS2 has been input to the T-STD).

[0441] The offset between STC1 and STC2 is determined as follows: the PTS1end is the PTS on STC1 corresponding to the last video presentation unit of TS2. PTS2start is PTS on STC2 corresponding to the first video presentation unit of TS2 and Tpp is the display time period of the last video presentation unit of TS1, the offset STC_delta between two system time bases is calculated in accordance with the following equation:

$$\text{STC_delta} = \text{PTS1end} + \text{Tpp} - \text{PTS2start}.$$

[0442] By way of explanation of the audio presentation timing, there may be overlap in the display timing of the audio presentation unit, with the overlap being less than 0 to 2 audio frames (see "audio overlap" shown in Fig. 97). The indication as to which of audio samples is to be selected and re-synchronization of the display of the audio presentation unit to the corrected time base at back of the junction point are set on the player.

[0443] By way of explaining the system time clock of DVR-STD, the last audio presentation unit of TS1 is displayed at time T5. The system time clock may be overlapped between time T2 and time T5. During this time domain, the DVR-STD switches the system time clocks between the value of the old time base (STC1) and the value of the new time base (STC2). The value of STC2 may be calculated in accordance with the following equation:

STC2 = STC1 - STC delta.

[0444] The buffering continuity is now explained. STC11video_end is the value of STC on the system time base STC2 when the first byte of the first video packet reaches TB1 of DVR-STD. STC22video_start is the value of STC on the system time base STC2 when the first byte of the first video packet reaches TB1 of DVR-STD. STC21video_end is the value of STC11video_end calculated as the value on STC2 of the system time base STC2. STC21video_end is calculated in accordance with the following equation:

STC21video.end = STC11video.end - STC.delta.

[0445] In order to obey DVR-STD, the following two conditions must be met: first, the arrival timing of the first video packet of TS2 at TB1 must satisfy the following inequality:

STC22video start > STC21video end + AT1.

[0446] If it is necessary to re-encode and/or multiplex the partial stream of Clip1 and/or Clip2 in such a manner that the above inequality will be satisfied, this re-encoding or multiplexing is performed as appropriate.

[0447] Second, the inputting of the video packet from TS1 followed by the inputting of the video packet from TS2 on the time axis of the system time base, mapped from STC1 and STC2 on

the same time axis, must not overflow or underflow the video buffer.

[0448] If the above syntax, data structure and the rules are used as basis, the contents of data recorded on the recording medium or the reproduction information can be managed properly to enable the user to confirm the contents of data recorded on the recording medium at the time of reproduction or to reproduce desired data extremely readily.

[0449] In the above-described embodiment, the MPEG 2 transport stream is taken as an example of the multiplexed stream. This, however, is merely exemplary, such that the MPEG 2 program stream DSS or the transport stream used in the DirecTV Service (trade mark) of USA may also be used as the multiplexed stream.

[0450] Fig. 98 shows a modification of BridgeSequenceInfo(). The difference from the BridgeSequenceInfo() of Fig. 38 is that it contains only Bridge_Clip_Information_file_name.

[0451] Fig. 99 illustrates Bridge_Clip when two PlayItems are seamlessly connected in case of using the syntax of BridgeSequenceInfo() of Fig. 98. RSPN_exit_from_previous_Clip is the source packet number of the source packet on the Clip AV stream referenced by the previous PlayItem. Next to this source packet is connected the first source packet of the Bridge_Clip AV stream file.

[0452] RSPN_enter_to_current_Clip is the number of the source packet on the Clip AV stream referenced by the current PlayItem. Ahead of this source packet is connected the last source packet of the Bridge-Clip AV stream file referenced by the current PlayItem. In the Bridge-Clip AV stream, shown in Fig. 99, SPN_ATC_start indicates the source packet number of the source

packet in the Bridge-Clip AV stream file at which commences the time axis of new arrival time base.

[0453] The Bridge-Clip AV stream file has a sole arrival time base discontinuous point. The second SPN_ATC_start in the drawings has the same meaning as that of the RSPN_arrival_time_discontinuity in Fig. 37.

[0454] If the syntax of BridgeSequenceInfo() of Fig. 98 is used, RSPN_exit_from_previous_Clip and RSPN_enter_to_current_Clip are stored in the Clip Information file corresponding to the Bridge-Clip AV stream file. SPN_ATC_start is also stored in the Clip Information file.

[0455] Fig. 100 shows the syntax of the Clip Information file when the BridgeSequenceInfo is of the syntax of Fig. 98. SequenceInfo_start_address indicates the leading address of SequenceInfo(), in terms of the number of relative bytes from the leading byte of the Clip Information file as a unit. The number of relative bytes is counted from 0.

[0456] Fig. 101 shows the syntax of ClipInfo() of the Clip Information file of Fig. 100. Clip_stream_type indicates whether the AV stream file of the Clip is the Clip AV stream file or the Bridge-Clip AV stream file. If the Clip_stream_type indicates the Bridge-Clip AV stream file, the next syntax field follows.

[0457] Previous_Clip_Information_file_name indicates the Clip Information file name connected ahead of the Bridge-Clip AV stream file. RSPN_exit_from_previous_Clip is the source packet number of the source packet on the Clip AV stream file indicated by previous_Clip_Information_file_name. Next to this source packet is connected the first source packet of the Bridge-Clip AV stream file. This source packet number is counted as from the

first source packet of the Clip AV stream file, with 0 as an initial value.

[0458] Current_Clip_Information_file_name indicates the Clip Information file name of the Clip connected at back of the Bridge-Clip AV stream file. RSPN_enter_to_current_Clip is the source packet number of the source packet on the Clip AV stream file indicated by the current_Clip_Information_file_name. Ahead of this source packet is connected the last source packet of the Bridge-Clip AV stream file. This source packet number is counted as from the first source packet of the Clip with the AV stream file with 0 as an initial value.

[0459] Fig. 102 indicates the syntax of the SequenceInfo() of the Clip Information file of Fig. 100. Num_of_ATC_sequences indicates the number of the ATC_sequence in the AV stream file. ATC_sequence is a source packet sequence not containing discontinuous points of the arrival time base. In the case of Bridge-Clip, this value is 2.

[0460] SPN_ATC_start[atc_id] indicates an address at which commences the arrival time base indicated by atc_id on the AV stream file. SPN_ATC_start[atc_id] is of a size based on the source packet number as a unit and is counted as from the first source packet of the AV stream file with 0 as an initial value.

[0461] Figs. 103A and 103B illustrate database change in case stream data of the Clip AV stream file referenced by Bridge-Sequence is partially erased. It is assumed that, as indicated by "Before Editing" in Fig. 103A, Clip1 and Clip2 are interconnected by Bridge-Clip, with RSPN_exit_from_previous_Clip = X and RSPN_enter_to_current_Clip = Y.

[0462] It is assumed that a stream data portion of Z1 source

packets of Clip1, shown shaded, and a stream data portion of Z1 source packets of Clip2, likewise shown shaded, are to be erased. As a result, RSPN_exit_from_previous_Clip becomes equal to X - Z1 and RSPN_enter_to_current_Clip becomes equal to Y - Z2, by way of value changing, as indicated by "After Editing" in Fig. 103B.

[0463] By changing the syntax of the database pertinent to BridgeSequence as shown in Figs. 98 and 101, the information pertinent to the source packet number indicating a data address in the AV stream (a field commencing from RSPN in the database syntax) is depleted from the layer of the PlayList, such that the totality of the information on the source packet number is stated in the Clip layer.

[0464] In this case, if it is necessary to change the value of the data address in the AV stream, as when data of the AV stream file has been partially erased, it is sufficient if merely the Clip Information file is supervised as data, thus advantageously facilitating database management.

[0465] Fig. 104 is a flowchart illustrating the preparation of the Real PlayList. Reference is had to the block diagram of Fig. 1 showing the recording and/or reproducing apparatus 1. At step S10, a controller 23 records a Clip AV stream. At step S11, the controller 23 forms PlayList() made up of PlayItems covering the totality of the possible playback range of the above Clip. If there is STC discontinuous point in the Clip, and PlayList() is made up of two or more PlayItems, the connection_condition between the PlayItems is also determined.

[0466] At step S12, the controller 23 forms UIAppInfoPlayList(). At step S13, the controller 23 forms PlayListMark. At step 14, the controller 23 forms

MakersPrivateData. At step S15, the controller 23 records a Real PlayList file. In this manner, one Real PlayList file is created each time a Clip AV stream file is newly recorded.

[0467] Fig. 105 is a flowchart for illustrating the formulation of a Virtual PlayList having a bridge sequence. At step S20, reproduction of one Real PlayList recorded on the disc is specified through a user interface. From the reproduction range of the Real PlayList, the reproduction domain specified by IN and OUT points is specified through a user interface.

[0468] At step S21, the controller 23 verifies whether or not the operation of specifying the reproduction range by the user has been finished in its entirety. If it is verified that the specifying operation has been finished, the controller 23 proceeds to step S22 and, if otherwise, the controller 23 reverts to step S20 to repeat the subsequent processing.

[0469] At step S22, the user determines the connection condition between two continuously reproduced PlayItems (connection_condition), through the user interface, or the controller 23 determines the connection condition. At step S23, the controller 23 forms a bridge sequence for the seamlessly connected PlayItems. At step S54, the controller 23 forms and records the Virtual PlayList file.

[0470] Fig. 106 is a flowchart for illustrating the detailed processing at step S23. At step S31, the controller 23 re-encodes and re-multiplexes an AV stream on the OUT point side of the PlayItem displayed on the temporally forward side. At step S32, the controller 23 re-encodes and re-multiplexes the AV stream on the IN point side of the PlayItem represented next to the PlayItem.

[0471] At step S33, the controller 23 determines the value of the RSPN_exit_from_previous_Clip such as to satisfy the data allocation condition for continuous data supply. That is, RSPN_exit_from_previous_Clip must be selected so that the stream portion of the Clip AV stream file previous to RSPN_exit_from_previous_Clip will be recorded in a continuous area not less than the aforementioned half fragment on the recording medium (see Figs. 91 and 94).

[0472] At step S34, the controller 23 determines the value of RSPN_enter_to_current_Clip such as to satisfy the data allocation condition for continuous data supply. That is, RSPN_enter_to_current_Clip must be selected so that the stream portion of the AV stream file subsequent to RSPN_enter_to_current_Clip will be arrayed in a continuous area not less than the aforementioned one fragment on the recording medium (see Figs. 91 and 94).

[0473] At step S35, the controller 23 forms the Bridge-Clip AV stream file such as to satisfy the data allocation condition for continuous data supply. That is, if the volume of data prepared in the processing of steps S31 and S32 is less than the size of the aforementioned one half fragment, data is copied from the original Clip to prepare Bridge-Clip (see Figs. 91 and 94).

[0474] Although the processing operations of steps S33 to S35 is explained chronologically, these processing operations may also be carried out non-sequentially or simultaneously since these processing operations are relevant reciprocally.

[0475] At step S36, the controller 23 forms a bridge sequence database. At step S37, the controller 23 records a Bridge-Clip AV stream file and its Clip information file. In this manner, one

or more PlayItem is selected by the user from the reproduction range of Real PlayList recorded on the disc. A bridge sequence for enabling seamless connection of two PlayItems is formed and a set of one or more PlayItems, grouped together, is recorded as one Virtual PlayList.

[0476] Fig. 107 is a flowchart for illustrating the reproduction of PlayList. At step S41, the controller 23 acquires Info.dvr, a Clip Information file, a PlayList file and the thumbnail information to prepare a GUI picture of a list of PlayLists recorded on the disc and to display the so formed picture on the GUI through the user interface.

[0477] At step S42, the controller 23 presents the information illustrating the PlayList on the GUI picture based on UIAppInfoPlayList() of each PlayList. At step S43, the user instructs reproduction of one PlayList from the GUI picture through a user interface. At step S44, the controller 23 acquires, from the PTS of IN_time and STC_sequence_id of the current PlayItem, a source packet having an entry point temporally previous and closest to IN_time.

[0478] At step S45, the controller 23 reads out data of the AV stream from the source packet number having the above entry point to send the so read out data to the decoder. If, at step S46, there is any PlayItem temporally previous to the current PlayItem, the controller 23 performs the process of interconnecting the previous PlayItem and the current PlayItem in accordance with connection_condition. If the PlayItems are interconnected seamlessly, the AV stream is decoded based on the DVR-STD decoding method.

[0479] At step S47, the controller 23 commands the AV decoder

27 to start the display as from the picture of the PTS of IN_time. At step S48, the controller 23 commands the AV decoder 27 to continue decoding the AV stream. At step S49, the controller 23 verifies whether or not the picture currently displayed is the picture of PTS of OUT_time. If it is verified that the picture currently displayed is not the picture of PTS of OUT_time, the controller 23 proceeds to step S50 to display the picture, after which the controller 23 reverts to step S48 to repeat the subsequent processing.

[0480] If it is verified at step S49 that the picture currently displayed is the picture of PTS of OUT_time, the controller 23 proceeds to step S51 where the controller 23 verifies whether or not the current PlayItem is the last one in the PlayList. If it is verified that the current PlayItem is not the last one, the controller 23 reverts to step S44 to repeat the subsequent processing. If it is verified that the current PlayItem is the last one, reproduction of the PlayList is finished.

[0481] In this manner, the one PlayList file, specified to be reproduced by the user, is reproduced.

[0482] Based on the above-described syntax, data structure and rule, as described above, the contents of data recorded on the recording medium, the playback information and so forth can be properly managed to enable the user to confirm the contents of data recorded on the recording medium properly to reproduce the desired data extremely readily.

[0483] The above-described sequence of operations may be executed not only by hardware but also by software. If the sequence of operations is to be executed by software, it is

installed from a recording medium to a computer in the dedicated hardware of which is installed the program forming the software or to, e.g., a general-purpose personal computer capable of executing various functions based on a variety of programs installed therein.

[0484] Fig. 108 shows an illustrative inner structure of a general-purpose computer. A CPU (central processing unit) 201 of the personal computer executes a variety of processing operations in accordance with a program stored in a ROM (read-only memory) 202. In a RAM (random-access memory) 203, there are suitably stored data or programs necessary for the CPU 201 to execute various processing operations. To an input/output interface 205 is connected an input unit 206 formed by a keyboard, a mouse and so forth to output signals input to the input unit 206 to the CPU 201. To the input/output interface 205, there is also connected an output unit 207 made up of a display, a loudspeaker and so forth.

[0485] To the input/output interface 205, there are also connected a recording unit 208 made up, e.g., of a hard disc and a communication unit 209 adapted for exchanging data with outer equipment over a network, such as the Internet. A drive 210 is used for reading out or writing data on or from a recording medium, such as a magnetic disc 221, an optical disc 222, a magneto-optical disc 223 or a semiconductor memory 224.

[0486] The recording medium is constituted not only by a package medium distributed for furnishing the program to the user, in addition to a computer, such as a magnetic disc 221 carrying the program thereon, inclusive of a floppy disc, an optical disc 222, inclusive of a CD-ROM (Compact Disc-Read-Only

memory) or a DVD (Digital Versatile Disc), a magneto-optical disc 223, inclusive of a Mini-Disc, or a semiconductor memory 224, but also by a hard disc, inclusive of a ROM 202 carrying a program and a memory 208, furnished to the user as it is built-in in a computer, as shown in Fig. 108.

[0487] In the present specification, the steps of the program furnished by the medium include not only the chronological processing in accordance with the sequence indicated, but also the processing performed not chronologically but in parallel or separately.

[0488] Additionally, in the specification, the system means an entire apparatus comprised of plural component devices.

[0489] In the information processing method and apparatus, and the program, according to the present invention, if the it is commanded to perform reproduction continuously from the first AV stream to the second AV stream, there is generated a third AV stream made up of a preset portion of the first AV stream and a preset portion of the second AV stream and which is reproduced when reproduction is switched from the first AV stream to the second AV stream, while there is generated, as information pertinent to the third AV stream, the address information made up of the information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and of the information on the address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream. So, reproduction may be achieved such as to maintain continuity in separately recorded AV streams.

[0490] In the information processing method and apparatus, and

the program, according to the present invention, a first AV stream, a second AV stream or a third AV stream is read out from a recording medium, the address information made up of the information on an address of a source packet of the first AV stream at a time of switching of reproduction from the first AV stream to the third AV stream, and the information on the address of a source packet of the second AV stream at a time of switching of reproduction from the third AV stream to the second AV stream, is read out from the recording medium as information pertinent to the third AV stream, and reproduction is switched from the first AV stream to the third AV stream and from the third AV stream to the second AV stream, as reproduction proceeds, based on the read-out information pertinent to the third AV stream. So, reproduction may be achieved such as to maintain continuity in separately recorded AV streams.

[0491] Although the invention herein has been described with reference to particular embodiments, it is to be understood that these embodiments are merely illustrative of the principles and applications of the present invention. It is therefore to be understood that numerous modifications may be made to the illustrative embodiments and that other arrangements may be devised without departing from the spirit and scope of the present invention as defined by the appended claims.